FACS 2011

8th International Symposium on Formal Aspects of Component Software

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Preliminary Proceedings
Editors: Farhad Arbab and Peter Csaba Ölveczky
Preface

This volume contains the preliminary proceedings of the 8th International Symposium on Formal Aspects of Component Software (FACS 2011), held at the Department of Informatics, University of Oslo, on September 14–16, 2011. FACS 2011 is the eighth event in a series of events founded by the International Institute for Software Technology of the United Nations University (UNU-IIST).

The objective of FACS is to bring researchers and practitioners of component software and formal methods together in order to foster a deeper understanding of reliable component-based systems development and their applications, using formal methods. The component-based software development approach has emerged as a promising paradigm to cope with the complexity of present-day software systems by bringing sound engineering principles into software engineering. However, many challenging conceptual and technological issues still remain in component-based software development theory and practice. Moreover, the advent of service-oriented computing has brought to the fore new dimensions, such as quality of service and robustness to withstand inevitable faults, that require revisiting established component-based concepts in order to meet the new requirements of the service-oriented paradigm.

FACS 2011 is concerned with how formal methods can be used to make component-based and service-oriented software development succeed. Formal methods have provided a foundation for component-based software by successfully addressing challenging issues such as mathematical models for components, composition and adaptation, or rigorous approaches to verification, deployment, testing, and certification. The symposium seeks to address the applications of formal methods in all aspects of software components and services.

In the call for papers for FACS 2011, we solicited high-quality submissions reporting on: original research contributions; applications and experiences; surveys, comparisons, and state-of-the-art reports; and tools. We also solicited submissions to the Doctoral Track of FACS 2011, in the form of two-page abstracts, concisely describing PhD-work-in-progress, conveying related theme, context, research questions, envisaged contributions, and partial results.

We received 46 submissions from 26 countries, out of which the program committee accepted 16 as regular papers, and, furthermore, conditionally accepted 4 additional papers. We received 9 submissions to the FACS 2011 Doctoral Track, out of which we accepted 4. Each submission to FACS 2011 was reviewed by at least 3 independent referees. In addition to the contributed papers, we consider ourselves very fortunate that our program also includes invited talks by José Meseguer, John Rushby, and Ketil Stølen.

Revised versions of accepted regular papers will appear in the post-proceedings of FACS 2011 that will be published as a volume in Springer’s LNCS series. Extended versions of selected papers from the symposium will also appear in a special issue of the Science of Computer Programming journal.

Many colleagues and friends have contributed to FACS 2011. First, we thank the authors who submitted their work to FACS 2011 and who, by their contributions and participation, make this symposium a high-quality event. We thank
the program committee members and their sub-reviewers for their timely and insightful reviews as well as for their involvement in the post-reviewing discussions. We are also grateful to the FACS Steering Committee for its guidance, to the invited speakers, and, in particular, to Lucian Bentea for all his assistance in organizing this event. Finally, we thank Andrei Voronkov for the excellent EasyChair conference system, which made our job so much simpler, and the Research Council of Norway and the Department of Informatics at the University of Oslo for financially supporting the symposium.

September, 2011

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Peter Csaba Olveczky
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Taming Distributed System Complexity through Formal Patterns

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Many current and future distributed systems are or will be:

- real-time and cyber-physical
- probabilistic in their operating environments and/or their algorithms
- safety-critical, with strong qualitative and quantitative formal requirements
- reflective and adaptive, to operate in changing and potentially hostile environments.

Their distributed features, their adaptation needs, and their real-time and probabilistic aspects make such systems quite complex and hard to design, build and verify, yet their safety-critical nature makes their verification essential. One important source of complexity, causing many unforeseen design errors, arises from ill-understood and hard-to-test interactions between their different distributed components.

Methods to tame and greatly reduce system complexity are badly needed. System complexity has many aspects, including the complexity and associated cost of:

- designing
- verifying
- developing
- maintaining and
- evolving

such systems.

The main goal of this talk is to propose the use of formal patterns as a way of drastically reducing all the above system complexity aspects. By a “formal pattern” I mean a solution to a commonly occurring problem that is:

- as generic as possible, with precise semantic requirements for its parameters
- formally specified
- executable, and
- comes with strong formal guarantees.

This means that a formal pattern can be applied to a potentially infinite set of concrete instances, where each such instance is correct by construction and enjoys the formal guarantees ensured by meeting the semantic requirements of the pattern’s parameters.

The overall vision is that distributed systems should be designed, verified, and built by composing formal patterns that are highly generic and reusable and
come with strong formal guarantees. A large part of the verification effort is spent in an up-front, fully generic manner, and is then be amortized across a potentially infinite number of instances. As I will show through concrete examples, this can achieve very drastic reductions in all aspects of system complexity, including the formal verification aspect. It can lead to high-quality, highly reliable distributed systems at a fraction of the cost required not using such patterns.

To develop formal patterns for distributed systems with features such as those mentioned above an appropriate semantic framework is needed, one supporting:

- concurrency
- real time and probabilities
- distributed reflection and adaptation, and
- formal verification methods.

I will use rewriting logic as such a semantic framework, and will show in number of examples its adequacy to specify and verify formal patterns of this nature.
Composing Safe Systems

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Abstract. We consider component-based systems that must ensure critical properties such as safety. We describe the value of partitioning, and of assumption synthesis, and suggest areas for further research.

1 Introduction

We build systems from components, but what makes something a system is that its properties and behavior are distinct from those of its components. As engineers and designers, we wish to predict and calculate the properties of systems from those of their components and their interconnections, and we are quite successful at doing this, most of the time. For many systems and properties, “most of the time” is good enough: we can live with it if our laptop occasionally locks up, our car doesn’t start, or our music player seems to lose our playlists. But we will be considerably more aggrieved if our laptop catches fire, our car fails to stop, or our music player loses the songs that we purchased. As we move from personal systems to those with wider impact and from properties about normal function to those that concern safety or security, so “most of the time” becomes inadequate: we want those properties to be true all the time.

Often, properties that we want to be true “all the time” fail to be so, and subsequent investigations generally reveal some unexpected interaction among the system’s components. Thus, attempts to reason about the properties of systems by combining or composing the properties of their components, while generally successful for “most of the time” properties, are less successful for “all the time” properties. It is for this reason that regulatory bodies examine only complete systems (e.g., the FAA certifies only airplanes and engines) and not components: they need to consider the behavior and possible interactions of multiple components in the context of a specific system.

Now, although it is generally infeasible at present to guarantee critical “all the time” properties by compositional (or “modular”) methods, it is a good research
topic to investigate why this is so, and how we might extend the boundaries of
what is feasible in this area. Safety, in the sense of causing no harm to the public,
is one of the most demanding properties, and so the motivation for the title of
this paper is to indicate a research agenda focused on methods that might allow
certification of safety for complex systems by compositional means.

As mentioned, system safety failures and the attendant flaws in compositional
reasoning are generally due to unanticipated interactions among components.
These interactions can be classified into those that exploit a previously unantici-
pated pathway for interaction, and those that are due to unanticipated behavior
along a known interaction pathway. One way to control the first class of unan-
ticipated interactions is to use integration frameworks that restrict the pathways
available for component interactions; in avionics, this approach is called “parti-
tioning” and it is the topic of Section 2.

There are two complementary ways to deal with the second class of unan-
ticipated interactions: one is to design components that deal gracefully with
anything their environment can do to them; the other is to figure out what
each component can deal with, and to ensure that its environment does not
subject it to anything outside that space. These ways of trying to “anticipate
the unanticipated” are among the topics of Section 3, which mainly focuses on
assume/guarantee and methods for assumption synthesis. Brief conclusions are
provided in Section 4.

2 Partitioning

Aircraft are safe, yet employ many interacting subsystems, so the techniques they
employ are worthy of interest. Traditionally, the various avionics functions on
board aircraft were provided by fairly independent subsystems loosely integrated
as a “federated” system. This meant that the autopilot, for example, had its own
computers, replicated for redundancy, and so did the flight management system.
The two system would communicate through the exchange of messages, but their
relative isolation provided a natural barrier to the propagation of faults: a faulty
autopilot might send bad data to the flight management system, but could not
 destroy its ability to calculate or communicate.

Modern aircraft employ Integrated Modular Avionics (IMA) where many
critical functions share the same computer system and communications net-
works, and so there is naturally concern that a fault in one function could prop-
gate to others sharing the same resources. Hence, the notion of “robust par-
titioning” has developed [11]: the idea is that applications that use the shared
resources should be protected from each other as if they were not sharing and
each had their own private resource.

The primary resources that require partitioning are communication and com-
putation: i.e., networks and processors. For networks, the concern is that a faulty
or malicious component will not adhere to the protocol and will transmit over
other traffic, or will transmit constantly, thereby denying good components the
ability to communicate. The only way to provide partitioning in the face of
these threats is to employ redundancy, so that components’ access to the network is mediated by additional components that limit the rate or the times at which communication can occur. Of course, these additional components and the mechanisms they employ may themselves be afflicted by faults (e.g., transient hardware upsets caused by ambient radiation), and so the design and assurance of these partitioning network technologies are very demanding [14], but they are now reasonably well understood and available “off the shelf.”

For processors, the concerns are that faulty or malicious processes will write into the memory of other processes, monopolize the CPUs, or corrupt the processor’s state. Partitioning against these threats can be provided by a strong operating system or, more credibly, by a hypervisor or virtual machine environment; minimal hypervisors specialized to the partitioning function are known as “separation kernels” [12] and, like partitioning networks, efficient and highly assured examples are now available “off the shelf.”

Partitioning for the basic resources of communication and computation can be leveraged to provide partitioning for additional resources synthesized above them, such as file systems. A collection of partitioning resources may be configured to specify quite precisely what software components are allocated to each partition and what interactions are allowed with other components (the configuration for an IMA is many megabytes of data). For some system-level properties, for example certain notions of security, such configurations, which may be portrayed as box and arrow diagrams and formalized as “policy architectures” [1], provide strong assurance.

![Fig. 1. A Partitioned System Configured to Support the System Purpose](image)

The properties for which partitioning, on its own, provides adequate enforcement and assurance are those that can be stated in terms of the absence of information flow. As mentioned, certain security concerns are of this kind (e.g., “no flow from secret to unclassified”), but most properties also concern the computations that take place in (some of) the partitions. For example, many secure systems do allow flow from secret to unclassified provided the information concerned is suitably “sanitized” by some function interposed in the flow, as portrayed by the minimal policy architecture of Figure 1. The partitioning configuration ensures the sanitizer cannot be bypassed, but we still require assurance that the sanitizer does its job. More complex properties, such as most notions of safety, cannot be ensured by individual components; instead, they emerge from the interactions of many—but partitioning eliminates unintended interactions and allows us to focus on correctness of the intended interactions, which is the topic of the next section.
3 Assumption Synthesis

If we suppose that “traditional software engineering” is able to develop systems that work “most of the time” then it might be possible to turn these into systems that work “all the time” by simply blocking the events or interactions that cause failures, or by controlling the failures that are precipitated. These topics are addressed by a variety of techniques such as systematic exception handling [4], anomaly detection [2], safety kernels [13] and enforceable security [18], and runtime monitoring [6]. All these techniques merely reduce the frequency or severity of failures (e.g., by turning “malfunction” or “unintended function” into “loss of function”) rather than eliminate them. However, they can be very valuable in systems with many layers of redundancy or fault management, since these often cope very well with the “clean” failure of subsystems, but less well with their misbehavior. Some aircraft systems employ “monitored architectures,” where a very simple component monitors the system safety property and shuts down the operational component if this is violated; these architectures can support rather strong assessments of reliability [9].

To get from clean failures to true “all the time” systems by compositional means, we need to be able to calculate the properties of the composed system from those of its components; if the calculation is automated, then it can support an iterative design loop: if the composed system does not satisfy the properties required, then we modify some of the components and their properties and repeat the calculation.

The established way to calculate the properties of interacting components is by assume/guarantee reasoning [7]: we verify that component A delivers (or guarantees) property p on the assumption that its environment delivers property q, and we also verify that B guarantees q assuming p; then when A and B are composed, each becoming the environment of the other, we may conclude (under various technical conditions) that their composition A||B guarantees both p and q. There is, however, a practical difficulty with this approach: if A and B are intended for general use, they are presumably developed in ignorance of each other, and it will require good fortune or unusual prescience that they should each make just the right assumption and guarantee that they fit together perfectly.

Shankar proposes an alternative approach [19] that treats assumptions as abstract components; here, we establish that p is a property of A in the context of an ideal environment E; if we can then show that B as a refinement of (i.e., a component with fewer behaviors than) E, then the composition of A and B also delivers p. This requires less prescience because we do not need to know about B at the time we design A; we do, however, need to postulate a suitable E.

One interesting idea is to design A, then calculate E as the weakest environment under which we can guarantee that A delivers p. When A is a concrete state machine, this can be done using $L^*$ learning [5]. Early in the design process, however, we are unlikely to have developed A to the point where it is available as a fully concrete design; in this case we can often perform assumption synthesis interactively using infinite bounded model checking (inf-BMC)
Inf-BMC performs bounded model checking on state machines defined over the theories supported by an SMT solver (i.e., a solver for Satisfiability Modulo Theories) \[15\]; these theories include equality over uninterpreted functions, possibly constrained by axioms, so it is possible to specify very abstract state machines. An example, taken from \[16\], is illustrated in Figure 2. Here, the goal is to deduce the assumptions under which a self-checking pair works correctly.

Self-checking pairs are used quite widely in safety-critical systems to provide protection against random hardware faults: two identical processors perform the same calculations and their results are compared; if they differ the pair shuts down (thereby becoming a “fail-stop” processor \[17\]) and some higher-level fault management activity takes over. Obviously, this does not work if both processors become faulty and compute the same wrong result. We would like to learn if there are any other scenarios that can cause a self-checking pair to deliver the wrong result; we can then assess their likelihood (for example, the double fault scenario just described may be considered extremely improbable) and calculate the overall reliability of this architecture.

The scenarios we wish to discover are, on one hand, the hazards to the design and, on the other (i.e., when negated), its assumptions. At the system level, hazard analysis is the cornerstone of safety engineering \[8\]; in component-based design, assumption discovery—its dual—could play a similar rôle: it helps us understand when it is safe for one component to become the environment for another.
Because of its context (the environment of a self-checking pair really is the natural environment, rather than another system), this example is closer to hazard discovery than assumption synthesis—but since these are two sides of the same coin, it serves to illustrate the technique. The idea is that the controller and the monitor are identical fault-prone computers that compute some uninterpreted function \( f(x) \); a distributor provides copies of the input \( x \) to both computers and the results are sent to a checker; if the results agree, the checker passes one of them on as safe out, otherwise it raises a fault flag. The distributor as well as the two computers can deliver incorrect outputs, but for simplicity of exposition the checker is assumed to be perfect (the checker can be eliminated by having the controller and monitor cross-check their results). An ideal computer, identical to the others but not subject to failures, serves as the correctness oracle, and an assumptions module, which operates as a synchronous observer, encodes the evolving assumptions. In the figure, the ideal and assumptions modules and their associated data are shown in red to emphasize that these are artifacts of analysis, not part of the component under design.

Initially, the assumptions are empty and we use inf-BMC to probe correctness of the design (i.e., we attempt to verify the claim that if the fault flag is down, then safe out equals ideal out). We obtain a counterexample that alerts us to a missing assumption; we add this assumption and iterate. The exercise terminates after the following assumptions are discovered.

1. When both members of the self-checking pair are faulty, their outputs should differ (this is the case we already thought of).
2. When the members of the pair receive different inputs\(^1\) (i.e., when the distributor is faulty), their outputs should differ. There are two subcases here.
   (a) Neither member of the pair is faulty. The scenario here is that instead of sending the correct value \( x \), the distributor sends \( y \) to one member of the pair and \( z \) to another, but \( f(y) = f(z) \) (and \( f(y) \neq f(x) \)).
   (b) One or both of the pair are faulty. Here, the scenario is the distributor sends the correct value \( x \) to the faulty member, and an incorrect value \( y \) to the nonfaulty member, but \( f(y) = f'(x) \), where \( f' \) is the computation of the faulty member.
3. When both members of the pair receive the same input, it is the correct input.

Inf-BMC can verify that the self-checking pair works, given these four assumptions, so our next task is to examine them.

Cases 1 and 2(b) require double faults and may be considered improbable. Case 2(a) is interesting because it probably would not be discovered by finite state model checking, where we do not have uninterpreted functions: instead,

\(^1\) Readers may wonder how a distributor, whose implementation could be as simple as a solder joint connecting two wires, can change values; one possibility is it adds resistance and drops the voltage: some receivers will see a weak voltage as a 1, and others as a 0.
the usual way to analyze an “abstract” design is to provide a very simple “concretization,” such as replacing $f(x)$ by $x+1$. This case is also interesting because, once discovered, it can be eliminated by modifying the design: simply cause each member of the pair to pass its input as well as its output to the checker; since both computers are nonfaulty, the inputs will be passed correctly to the checker, which will then rise the fault flag because it sees that the inputs differ. That leaves case 3 as the one requiring further consideration (which we do not pursue here) by those who would use a self-checking pair.

This example has illustrated, I hope, how automated methods such as inf-BMC can be used to help calculate the weakest assumptions required by a component, and thereby support the design of systems in which components’ assumes and guarantees are mutually supportive, without requiring prescience.

4 Conclusions

All fields of engineering build on components, and it is natural that computer science should do the same. However, component-based systems can be rather more challenging in computer science than in other fields because of the complexity of interaction—unintended as well as in intended—that is possible. This complexity of interaction becomes even more vexatious when we aim to develop safety-critical and other kinds of system that must work correctly all the time. (Perrow [10] would argue that unintended interactions and their enablers, “interactive complexity” and “tight coupling,” are the primary causes of disasters in all engineering fields; however, computer systems generally have more complexity of these kinds, even in normal operation, than those of other fields.)

Unintended interactions can be divided into those that deliver unintended behavior along intended pathways, and those that employ unintended pathways. I have outlined techniques that can ameliorate these concerns. Partitioning aims to eliminate unintended pathways for interaction in networks and processors and higher-level resources built on these. Partitioning guarantees “preservation of prior properties” when new components are added to an existing system; it also seems sufficient, on its own, to guarantee certain kinds of information flow security properties, and to simplify the assured construction of more complex properties of this kind [3]. With unintended pathways controlled by partitioning, we can turn to interactions along known pathways. Various techniques akin to wrapping and monitoring can reduce bad inputs and outputs, but ultimately we need to calculate the composed behavior of interacting components. Traditional methods of assume/guarantee reasoning demand a degree of prescience to ensure that the assumes of one component are met by the guarantees of another, designed in ignorance of it. One way to lessen this need for prescience is to derive the weakest assumptions under which a component can deliver its guarantees, and I sketched how inf-BMC can be used to help automate this process (which is closely related to hazard analysis) very early in the design cycle.

Compositional design and assurance for critical systems that must function correctly, or at least safely, all the time, are challenging and attractive research
Further systematic examination and study of the methods and directions I described could be worthwhile, but fresh thinking would also be welcome.

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Components and Risk

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Abstract. Risk analysis is an important tool to establish the appropriate protection or safety level of a system. Unfortunately, the shifting environment of components is not adequately addressed by traditional risk analysis methods. Furthermore, the issue of risk is hardly addressed within methods for component-based system development. This talk will present challenges with respect to components and risk from three viewpoints, namely an industrial viewpoint, a modelling viewpoint and a theoretical viewpoint. The presentation builds on the following publications [3], [1] and [2].

References

Synthesis of Hierarchical Systems

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Abstract In automated synthesis, given a specification, we automatically create a system that is guaranteed to satisfy the specification. In the classical temporal synthesis algorithms, one usually creates a “flat” system “from scratch”. However, real-life software and hardware systems are usually created using pre-existing libraries of reusable components, and are not “flat” since repeated subsystems are described only once.

In this work we describe an algorithm for the synthesis of a hierarchical system from a library of hierarchical components, which follows the “bottom-up” approach to system design. Our algorithm works by synthesizing in many rounds, when at each round the system designer provides the specification of the currently desired module, which is then automatically synthesized using the initial library and the previously constructed modules. To ensure that the synthesized module actually takes advantage of the available high-level modules, we guide the algorithm by enforcing certain modularity criteria.

We show that the synthesis of a hierarchical system from a library of hierarchical components is $\text{EXPTIME}$-complete for $\mu$-calculus, and $2\text{EXPTIME}$-complete for $\text{LTL}$, both in the cases of complete and incomplete information. Thus, in all cases, it is not harder than the classical synthesis problem (of synthesizing flat systems “from scratch”), even though the synthesized hierarchical system may be exponentially smaller than a flat one.

1 Introduction

Synthesis is the automated construction of a system from its specification. The basic idea is simple and appealing: instead of developing a system and verifying that it is correct w.r.t. its specification, we use instead an automated procedure that, given a specification, constructs a system that is correct by construction. The first formulation of synthesis goes back to Church [7]; the modern approach to this problem was initiated by Pnueli and Rosner who introduced linear temporal logic (LTL) synthesis [23], later extended to handle branching-time specifications, such as $\mu$-calculus [10].

The Pnueli and Rosner idea can be summarized as follows. Given sets $\Sigma_I$ and $\Sigma_O$ of inputs and outputs, respectively (usually, $\Sigma_I = 2^I$ and $\Sigma_O = 2^O$, where $I$ is a set of input signals and $O$ is a set of output signals), we can view a system as a strategy $P : \Sigma_I \to \Sigma_O$ that maps a finite sequence of sets of input signals into a set of output signals. When $P$ interacts with an environment that generates infinite input sequences, it associates with each input sequence an infinite computation over $\Sigma_I \cup \Sigma_O$. Though the system $P$ is deterministic, it induces a computation tree. The branches of the tree

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correspond to external nondeterminism, caused by different possible inputs. Thus, the
tree has a fixed branching degree $|\Sigma_I|$, and it embodies all the possible inputs (and
hence also computations) of $P$. When we synthesize $P$ from an LTL specification $\varphi$, we
require $\varphi$ to hold in all the paths of $P$'s computation tree. However, in order to impose
possibility requirements on $P$, we have to use a branching-time logic like $\mu$-calculus.
Given a branching specification $\varphi$ over $\Sigma_I \cup \Sigma_O$, realizability of $\varphi$ is the problem of
determining whether there exists a system $P$ whose computation tree satisfies $\varphi$. Correct
synthesis of $\varphi$ then amounts to constructing such a $P$.

In spite of the rich theory developed for system synthesis in the last two decades,
little of this theory has been reduced to practice. In fact, the main approaches to tackle
synthesis in practice are either to use heuristics (e.g., [13]) or to restrict to simple spec-
fications (e.g., [22]). Some people argue that this is because the synthesis problem is
very expensive compared to model-checking [16]. There is, however, something mis-
leading in this perception: while the complexity of synthesis is given with respect to
the specification only, the complexity of model-checking is given also with respect to a
program, which can be very large. A common thread in almost all of the works concern-
ing synthesis is the assumption that the system is to be built “from scratch”. Obviously,
real-world systems are rarely constructed this way, but rather by utilizing many pre-
existing reusable components, i.e., a library. Using standard preexisting components is
sometimes unavoidable (for example, access to hardware resources is usually under the
control of the operating system, which must be “reused”), and many times has other
benefits (apart from saving time and effort, which may seem to be less of a problem
in a setting of automatic - as opposed to manual - synthesis), such as maintaining a
common code base, and abstracting away low level details that are already handled by
the preexisting components. Another reason that may also account, at least partially, for
the limited use of synthesis in practice, is the fact that synthesized systems are usually
monolithic and look very unnatural from the system designer’s point of view. Indeed,
in classical temporal synthesis algorithms, one usually creates a “flat” system, i.e., a
system in which sub-systems may be repeated many times. On the contrary, real-life
software and hardware systems are hierarchical (or even recursive) and repeated sub-
systems (such as sub-routines) are described only once. While hierarchical systems may
be exponentially more succinct than flat ones, it has been shown that the cost of solv-
ing questions about them (like model-checking) are in many cases not exponentially
higher [5, 6, 12]. Hierarchical systems can also be seen as a special case of recursive
systems [2, 3], where the nesting of calls to sub-systems is bounded. However, having
no bound on the nesting of calls gives rise to infinite-state systems, and this results in a
higher complexity.

In this work we provide a uniform algorithm, for different temporal logics, for the
synthesis of hierarchical systems (or, equivalently, transducers) from a library of hi-
erarchical systems, which mimics the “bottom-up” approach to system design, where
one builds a system by iteratively constructing new modules based on previously con-
structed ones\(^1\). More specifically, we start the synthesis process by providing the algo-

\(^1\) While for systems built from scratch, a top-down approach may be argued to be more suitable,
we find the bottom-up approach to be more natural when synthesizing from a library.
which is then automatically synthesized using the currently available components as possible sub-components. Once a new component is synthesized, it is added to the library to be used by subsequent iterations. We show that while hierarchical systems may be exponentially smaller than flat ones, the problem of synthesizing a hierarchical system from a library of existing hierarchical systems is \text{EXPTIME}-complete for \( \mu \)-calculus, and \( 2\text{EXPTIME} \)-complete for \( LTL \). Thus, this problem is not harder than the classical synthesis problem of flat systems “from scratch”. Furthermore, we show that this is true also in the case where the synthesized system has incomplete information about the environment’s input.

Observe that it is easily conceivable that if the initial library \( L_0 \) contains enough atomic components then the synthesis algorithm may use them exclusively, essentially producing a flat system. We thus have to direct the single-round synthesis algorithm in such a way that it produces modular and not flat results. The question of what makes a design more or less modular is very difficult to answer, and has received many (and often widely different) answers throughout the years (see [21], for a survey). We claim that some very natural modularity criteria are regular, and show how any criterion that can be checked by a parity tree automaton can be easily incorporated into our automata based synthesis algorithm.

Related work The issues of specification and correctness of modularly designed systems have received a fair attention in the formal verification literature. Examples of important work on this subject are [8, 17, 26, 27]. On the other hand, the problem of automatic synthesis from reusable components, which we study here, has received much less attention. The closest to our work is Lustig and Vardi’s work on \( LTL \) synthesis from libraries of (flat) transducers [18]. The technically most difficult part of our work is an algorithm for performing the synthesis step of a single round of the multiple-rounds algorithm. To this end, we use an automata-theoretic approach. However, unlike the classical approach of [23], we build an automaton whose input is not a computation tree, but rather a system description in the form of a connectivity tree (inspired by the “control-flow” trees of [18]), which describes how to connect library components in a way that satisfies the specification formula. Taken by itself, our single-round algorithm extends the “control-flow” synthesis work from [18] in four directions. (i) We consider not only \( LTL \) specifications but also the modal \( \mu \)-calculus. Hence, unlike [18], where co-Büchi tree automata were used, we have to use the more expressive parity tree automata. Unfortunately, this is not simply a matter of changing the acceptance condition. Indeed, in order to obtain an optimal upper bound, a widely different approach, which makes use of the machinery developed in [6] is needed. (ii) We need to be able to handle libraries of hierarchical transducers, whereas in [18] only libraries of flat transducers are considered. (iii) A synthesized transducer has no top-level exits (since it must be able to run on all possible input words), and thus, its ability to serve as a sub-transducer of another transducer (in future iterations of the multiple-rounds algorithm) is severely limited (it is like a function that never returns to its caller). We therefore need to address the problem of synthesizing exits for such transducers. (iv) As discussed above, we incorporate into the algorithm the enforcing of modularity criteria.

Recently, an extension of [18] appeared in [19], where the problem of Nested-Words Temporal Logic (NWTL) synthesis from recursive component libraries has been investigated. NWTL extends \( LTL \) with special operators that allow one to handle “call and return” computations [1] and it is used in [19] to describe how the components have to be connected in the synthesis problem. We recall that in our framework the logic
does not drive (at least not explicitly) the way the components have to be connected. Moreover, the approach used in [19] cannot be applied directly to the branching framework we consider in this paper, as we recall that already the satisfiability problem for \( \mu \)-calculus with “call and return” is undecidable even for very restricted cases [4].

Due to lack of space some proofs are omitted and reported in a full version found at the authors’ web page.

2 Alternating Tree Automata

Let \( D \) be a set. A \( D \)-tree is a prefix-closed subset \( T \subseteq D^* \) such that if \( x \cdot c \in T \), where \( x \in D^* \) and \( c \in D \), then also \( x \in T \). The complete \( D \)-tree is the tree \( D^* \). The elements of \( T \) are called \textit{nodes}, and the empty word \( \varepsilon \) is the \textit{root} of \( T \). Given a word \( x = y \cdot d \), with \( y \in D^* \) and \( d \in D \), we define \textit{last} \( (x) \) to be \( d \). For \( x \in T \), the nodes \( x \cdot d \in T \), where \( d \in D \), are the \textit{sons} of \( x \). A \textit{leaf} is a node with no sons. A \textit{path} of \( T \) is a set \( \pi \subseteq T \) such that \( \varepsilon \in \pi \) and, for every \( x \in \pi \), either \( x \) is a leaf or there is a unique \( d \in D \) such that \( x \cdot d \in \pi \). For an alphabet \( \Sigma \), a \textit{\( \Sigma \)-labeled} \( D \)-tree is a pair \((T, V)\) where \( T \subseteq D^* \) is a \( D \)-tree and \( V : T \rightarrow \Sigma \) maps each node of \( T \) to a letter in \( \Sigma \).

\textit{Alternating tree automata} are a generalization of nondeterministic tree automata [20] (see [16], for more details). Intuitively, while a nondeterministic tree automaton visits a node of the input tree sends exactly one copy of itself to each of the sons of the node, an alternating automaton can send several copies of itself to the same son.

An (asymmetric) \textit{Alternating Parity Tree Automaton (APT)} is a tuple \( A = (\Sigma, D, Q, q_0, \delta, F) \), where \( \Sigma, D, \) and \( Q \) are non-empty finite sets of \textit{input letters}, \textit{directions}, and \textit{states}, respectively; \( q_0 \in Q \) is an \textit{initial state}, \( F \) is a \textit{parity acceptance condition} to be defined later, and \( \delta : Q \times \Sigma \rightarrow B^* (D \times Q) \) is an \textit{alternating transition function}, which maps a state and an input letter to a positive boolean combination of elements in \( D \times Q \).

Given a set \( S \subseteq D \times Q \) and a formula \( \theta \in B^* (D \times Q) \), we say that \( S \) \textit{satisfies} \( \theta \) (denoted by \( S \models \theta \)) if assigning \textit{true} to elements in \( S \) and \textit{false} to elements in \((D \times Q) \setminus S\), makes \( \theta \) true. A \textit{run} of an APT \( A \) on a \( \Sigma \)-labeled \( D \)-tree \( T = (T, V) \) is a \((T \times Q)\)-labeled \( \Sigma \)-tree \((T, r)\), where \( \Sigma = (S \cup \{\varepsilon\}) \), the set of \textit{non-negative integers}, such that (i) \( r(\varepsilon) = (\varepsilon, q_0) \) and (ii) for all \( x \in T \), with \( r(y) = (x, q) \), there exists a set \( S \subseteq D \times Q \), such that \( S \models \delta(q, V(x)) \), and there is one son \( y' \) of \( y \), with \( r(y') = (x \cdot d, q') \), for every \( (d, q') \in S \). Given a node \( x \) of a run \((T, r)\), with \( r(x) = (\varepsilon, q) \in T \times Q \), we define \textit{last} \( (r(y)) = (\text{last}(z), q) \). An alternating parity automaton \( A \) is nondeterministic (denoted NPT), iff when its transition relation is rewritten in disjunctive normal form each disjunct contains at most one element of \( \{d\} \times Q \), for every \( d \in D \). An automaton is universal (denoted UPT) if all the formulas that appear in its transition relation are conjunctions of atoms in \( D \times Q \).

A symmetric alternating parity tree automaton with \( \varepsilon \)-moves (SAPT) [14] does not distinguish between the different sons of a node, and can send copies of itself only in a universal or an existential manner. Formally, an SAPT is a tuple \( A = (\Sigma, Q, q_0, \delta, F) \), where \( \Sigma \) is a finite input alphabet; \( Q \) is a finite set of states, partitioned into universal \((Q^U)\), existential \((Q^E)\), and \( \varepsilon \)-and \((Q^{\varepsilon\wedge})\), and \( \varepsilon \)-or \((Q^{\varepsilon\lor})\) states (we also write \( Q^\wedge = Q^U \cup Q^E \), and \( Q^\lor = Q^{\varepsilon\wedge} \cup Q^{\varepsilon\lor} \)); \( q_0 \in Q \) is an initial state; \( \delta : Q \times \Sigma \rightarrow (Q \cup 2^Q) \) is a transition function such that for all \( \sigma \in \Sigma \), we have that \( \delta(q, \sigma) \in Q \) for \( q \in Q^\wedge \), and \( \delta(q, \sigma) \in 2^Q \) for \( q \in Q^\lor \); and \( F \) is a parity acceptance condition, to be defined later. We assume that \( Q \) contains in addition two special states \( \varepsilon \) and \( tt \), called \textit{rejecting sink} and \textit{accepting sink}, respectively, such that \( \forall a \in \Sigma : \delta(tt, a) = tt, \delta(\varepsilon, a) = \varepsilon. \) The classification of \( tt \) and \( tt \) is arbitrary. Transitions from states in \( Q^E \) launch copies of \( A \)
that stay on the same input node as before the transition, while transitions from states in $Q^\lor$ launch copies that advance to sons of the current node (note that for an SAPT the set $D$ of directions of the input trees plays no role in the definition of a run). When a symmetric alternating tree automaton $A$ runs on an input tree it starts with a copy in state $q_0$ whose reading head points to the root of the tree. It then follows $\delta$ in order to send further copies. For example, if a copy of $A$ that is in state $q \in Q^\ell$ is reading a node $x$ labeled $\sigma$, and $\delta(q, \sigma) = \{q_1, q_2\}$, then this copy proceeds either to state $q_1$ or to state $q_2$, and its reading head stays in $x$. As another example, if $q \in Q^\lor$ and $\delta(q, \sigma) = q_1$, then $A$ sends a copy in state $q_1$ to every son of $x$. Note that different copies of $A$ may have their reading head pointing to the same node of the input tree. Formally, a run of $A$ on a $\Sigma$-labeled $D$-tree $(T, V)$ is a $(T \times Q)$-labeled $I$-tree $(T, r)$. A node in $T$, labeled by $(x, q)$ describes a copy of $A$ in state $q$ that reads the node $x$ of $T$. A run has to satisfy $r(x) = (x, q_0)$ and, for all $y \in T$, with $r(y) = (x, q)$, the following hold:

- If $q \in Q^\ell$ (resp. $q \in Q^\lor$) and $\delta(q, V(x)) = p$, then for each son (resp. for exactly one son) $x \cdot d$ of $x$, there is a node $y \cdot i \in T$, with $r(y \cdot i) = (x \cdot d, p)$.
- If $q \in Q^{(\ell \lor)}$ (resp. $q \in Q^{(\lor \ell)}$) and $\delta(q, V(x)) = \{p_0, \ldots, p_k\}$, then for all $i \in \{0, \ldots, k\}$ (resp. for one $i \in \{0, \ldots, k\}$) the node $y \cdot i \in T$, and $r(y \cdot i) = (x, p_i)$.

A parity condition is given by means of a coloring function on the set of states. Formally, a parity condition is a function $F : Q \rightarrow C$, where $C = \{C_{\min}, \ldots, C_{\max}\} \subseteq \mathbb{N}$ is a set of colors. The size $|C|$ of $C$ is called the index of the automaton. For an SAPT, we also assume that the special state $\top$ is given an even color, and $\ell \top$ is given an odd color. For an infinite path $\pi \subseteq T$, of a run $(T, r)$, let $\max C(\pi)$ be the maximal color that appears infinitely often along $\pi$. Similarly, for a finite path $\pi$, we define $\max C(\pi)$ to be the maximal color that appears at least once in $\pi$. An infinite path $\pi \subseteq T$ satisfies the acceptance condition $F$ iff $\max C(\pi)$ is even. A run $(T, r)$ satisfies $\ell F$ all its infinite paths satisfy $F$. The automaton $A$ accepts an input tree $(T, V)$ if there is an accepting run of $A$ on $(T, V)$. The language of $A$, denoted $L(A)$, is the set of $\Sigma$-labeled $D$-trees accepted by $A$. We say that an automaton $A$ is nonempty iff $L(A) \neq \emptyset$.

A wide range of branching-time temporal logics can be translated to alternating tree automata (details can be found in [16]). In particular:

**Theorem 1.** [11, 16] Given a temporal-logic formula $\varphi$, it is possible to construct a SAPT $A_\varphi$ such that $L(A_\varphi)$ is exactly the set of trees satisfying $\varphi$. Moreover, (i) if $\varphi$ is a $\mu$-calculus formula, then $A_\varphi$ is an alternating parity automaton with $O(|\varphi|)$ states and index $O(|\varphi|)$; and (ii) if $\varphi$ is an LTL formula, then $A_\varphi$ is a universal parity automaton with $2\Theta(|\varphi|)$ states, and index 2.

### 3 Hierarchical Transducers

In this section, we introduce hierarchical transducers (alternatively, hierarchical Moore machines), which are a generalization of classical transducers in which repeated substructures (technically, sub-transducers) are specified only once. Technically, some of the states in a hierarchical transducer are boxes, in which inner hierarchical transducers are nested. Formally, a hierarchical transducer is a tuple $\mathcal{K} = \langle \Sigma_I, \Sigma_O, \{\mathcal{K}_i, \ldots, \mathcal{K}_n\} \rangle$, where $\Sigma_I$ and $\Sigma_O$ are respectively non-empty sets of input and output letters, and for every $1 \leq i \leq n$, the sub-transducer $\mathcal{K}_i = \langle W_i, B_i, \text{ini}_i, \text{Exit}_i, \tau_i, \delta_i, \Lambda_i \rangle$ has the following elements.
- $W_i$ is a finite set of states. $i_0 \in W_i$ is an initial state, and $\text{Exit}_i \subseteq W_i$ is a set of exit-states. States in $W_i \setminus \text{Exit}_i$ are called internal states.
- A finite set $B_i$ of boxes. We assume that $W_1, \ldots, W_n, B_1, \ldots, B_k$ are pairwise disjoint.
- An indexing function $\tau : B_i \rightarrow \{i + 1, \ldots, n\}$ that maps each box of the $i$-th sub-transducer to a sub-transducer with an index greater than $i$. If $\tau(b) = j$ we say that $b$ refers to $K_j$.
- A transition function $\delta_i : (\cup_{b \in B_i} \{b\} \times \text{Exit}_{\tau(b)}) \cup (W_i \setminus \text{Exit}_i) \times \Sigma \rightarrow W_i \cup B_i$.
- A labeling function $\Lambda_i : W_i \rightarrow \Sigma_O$ that maps states to output letters.

The sub-transducer $K_i$ is called the top-level sub-transducer of $K$. Thus, for example, the top-level boxes of $K$ are the elements of $B_1$, etc. We also call $i_0$ the initial state of $K$, and $\text{Exit}_1$ the exits of $K$. For technical convenience we sometimes refer to functions (like the transitions and labeling functions) as relations, and in particular, we consider $\emptyset$ to be a function with an empty domain. Note that the fact that boxes can refer to sub-transducers of a greater index implies that the nesting depth of transducers is finite. In contrast, in the recursive setting such a restriction does not exist. Also note that moves from an exit $e \in \text{Exit}_i$ of a sub-transducer $K_i$ are not specified by the transition function $\delta_i$ of $K_i$, but rather by the transition functions of the sub-transducers that contain boxes that refer to $K_i$. The exits of $K_i$ allow us to use it as a sub-transducer of another hierarchical transducer. When we say that a hierarchical transducer $K = \langle \Sigma_i, \Sigma_o, (K_1, \ldots, K_n) \rangle$ is a sub-transducer of another hierarchical transducer $K' = \langle \Sigma_i, \Sigma_o, (K'_1, \ldots, K'_n) \rangle$, we mean that $\{K_1, \ldots, K_n\} \subseteq \{K'_1, \ldots, K'_n\}$. The size $|K|$ of a sub-transducer $K$ is the sum $|W_i| + |B_i| + |\delta_i|$. The size $|K|$ of $K$ is the sum of the sizes of its sub-transducers. We sometimes abuse notation and refer to the hierarchical transducer $K_i$ which is formally the hierarchical transducer $\langle \Sigma_i, \Sigma_o, (K_i, K_{i+1}, \ldots, K_n) \rangle$ obtained by taking $K_i$ to be the top-level sub-transducer.

**Flat transducers** A sub-transducer without boxes is flat. A hierarchical transducer $K = \langle \Sigma_i, \Sigma_o, \langle W, \emptyset, \text{in}, \text{Exit}, \emptyset, \delta, \Lambda \rangle \rangle$ with a single (hence flat) sub-transducer is flat, and we denote it using the shorter notation $K = \langle \Sigma_i, \Sigma_o, \langle W, \text{in}, \text{Exit}, \delta, \Lambda \rangle \rangle$. Each hierarchical transducer $K$ can be transformed into an equivalent flat transducer $K^f = \langle \Sigma_i, \Sigma_o, \langle W^f, \text{in}, \text{Exit}, \delta^f, \Lambda^f \rangle \rangle$ (called its flat expansion) by recursively substituting each box by a copy of the sub-transducer it refers to. Since different boxes can refer to the same sub-transducer, states may appear in different contexts. In order to obtain unique names for states in the flat expansion, we prefix each copy of a sub-transducer’s state by the sequence of boxes through which it is reached. Thus, a state $(b_0, \ldots, b_k, w)$ of $K^f$ is a vector whose last component $w$ is a state in $\cup_{i=1}^{k} W_i$, and the remaining components $(b_0, \ldots, b_k)$ are boxes that describe its context. The labeling of a state $(b_0, \ldots, b_k, w)$ is determined by its last component $w$. For simplicity, we refer to vectors of length one as elements (that is, $w$, rather than $(w)$).

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2 We assume a single entry for each sub-transducer. Multiple entries can be handled by duplicating sub-transducers.

3 A helpful way to think about this is using a stack — the boxes $b_0, \ldots, b_k$ are pushed into the stack whenever a sub-transducer is called, and are popped in the corresponding exit.
The set of states $W_f \subseteq W \cup (B \times (\bigcup_{j=1}^{f} W_j))$ is defined as follows: (i) if $w$ is a state of $W_l$ then $w$ belongs to $W_f$; and (ii) if $b$ is a box of $K$, with $\tau_i(b) = j$, and the tuple $(u_1, \ldots, u_h)$ is a state in $W_j$, then $(b, u_1, \ldots, u_h)$ belongs to $W_f$.

The transducer $\langle \Sigma, \Sigma, (K_i) \rangle$ is the required flat expansion $K_f = \langle W_f, \text{init}, \text{Exit}_f, \tau_f, \delta_f, \Lambda_f \rangle$ as follows.

- The transition function $\delta_f$ is defined as follows: (i) If $\delta_f(u, \sigma) = v$, where $u \in W$, or $u = (b, e)$ with $b \in B$ and $e \in \text{Exit}_f(b)$, then if $v$ is a state, we have that $\delta_f(u, \sigma) = v$; and if $v$ is a box, we have that $\delta_f(u, \sigma) = (v, \text{init}(v))$. Note that $(v, \text{init}(v))$ is indeed a state of $W_f$ by the second item in the definition of states above; and (ii) if $b$ is a box of $K$, and $\delta_f(b)(u_1, \ldots, u_h, \sigma) = (v_1, \ldots, v_h')$ is a transition of $K$, then $\delta_f((b, u_1, \ldots, u_h), \sigma) = (b, v_1, \ldots, v_h')$ is a transition of $K_f$.

- Finally, if $u \in W_l$ then $\Lambda_f(u) = \Lambda_f(u)$; and if $u \in W_f$ is of the form $u = (b, u_1, \ldots, u_h)$, where $b \in B$, then $\Lambda_f(u) = \Lambda_f(b)(u_1, \ldots, u_h)$.

The transducer $\langle \Sigma, \Sigma, (K_f) \rangle$ is the required flat expansion $K_f$ of $K$. An atomic transducer is a flat transducer made up of a single node that serves as both an entry and an exit. For each letter $\xi \in \Sigma$ there is an atomic transducer $K_\xi = \{ (p, p, \emptyset, \emptyset) \}$ whose single state $p$ is labeled by $\xi$.

**Run of a hierarchical transducer** Consider a hierarchical transducer $K$ with $\text{Exit}_f = \emptyset$ that interacts with its environment. At point $j$ in time, the environment provides $K$ with an input $\sigma_j \in \Sigma$, and in response $K$ moves to a new state, according to its transition relation, and outputs the label of that state. The result of this infinite interaction is a computation of $K$, called the *trace* of the run of $K$ on the word $\sigma_1 \cdot \sigma_2 \cdots$. In the case that $\text{Exit}_f \neq \emptyset$, the interaction comes to a halt whenever $K$ reaches an exit $e \in \text{Exit}_f$, since top-level exits have no outgoing transitions. Formally, a *run* of a hierarchical transducer $K$ is defined by means of its flat expansion $K_f$. Given a finite input word $v = \sigma_1 \cdots \sigma_m \in \Sigma^*$, a run (computation) of $K$ on $v$ is a sequence of states $r = r_0 \cdots r_m \in (W_f)^*$ such that $r_0 = \text{init}_1$, and $r_j = \delta_f(r_{j-1}, \sigma_j)$, for all $0 < j \leq m$. Note that since $K_f$ is deterministic it has at most one run on every word, and that if $\text{Exit}_f \neq \emptyset$ then $K$ may not have a run on some words. The *trace* of the run of $K$ on $v$ is the word of inputs and outputs $\text{trc}(K, v) = (\Lambda_f(r_1), \sigma_1) \cdot (\Lambda_f(r_m), \sigma_m) \in (\Sigma \times \Sigma)^*$. The notions of traces and runs are extended to infinite words in the natural way.

The computations of $K$ can be described by a *computation tree* whose branches correspond to the runs of $K$ on all possible inputs, and whose labeling gives the traces of these runs. Note that the root of the tree corresponds to the empty word $\epsilon$, and its labeling is not part of any trace. However, if we look at the computation tree of $K$ as a sub-tree of a computation tree of a transducer $K'$ of which $K$ is a sub-transducer, then the labeling of the root of the computation tree of $K'$ is meaningful, and it corresponds to the last element in the trace of the run of $K'$ leading to the initial state of $K$. Formally, given $\sigma \in \Sigma$, the computation tree $T_{K, \sigma} = \langle \Sigma \times \Sigma, V_{K, \sigma} \rangle$, is a $(\Sigma \times \Sigma)$-labeled $(\Sigma \times \Sigma)$-tree, where: (i) the root $\epsilon$ is labeled by $(\Lambda_f(\text{init}_1), \sigma)$; (ii) a node $y = (r_1, \sigma_1) \cdots (r_m, \sigma_m) \in (W_f \times \Sigma)^*$ is in $T_{K, \sigma}$ iff $\text{init}_1 \cdots r_1 \cdots r_m$ is the run of $K$ on $v = \sigma_1 \cdots \sigma_m$, and its label is $V_{K, \sigma}(y) = (\Lambda_f(r_m), \sigma_m)$. Thus, for a node $y$, the labels of the nodes on the path from the root (excluding the root) to $y$ are exactly $\text{trc}(K, v)$. Observe that the leaves of $T_{K, \sigma}$ correspond to pairs $(e, \sigma')$, where $e \in \text{Exit}_f$ and $\sigma' \in \Sigma$,
However, if \( \text{Exit}_1 = /0 \), then the tree has no leaves, and it represents the runs of \( K \) over all words in \( \Sigma^* \). We sometimes consider a leaner computation tree \( T_K = \langle T_K, V_K \rangle \) that is a \( \Sigma_O \)-labeled \( \Sigma_I \)-tree, where a node \( y \in \Sigma_I^* \) is in \( T_K \) iff there is a run \( r \) of \( K \) on \( y \). The label of such a node is \( V_K(y) = \Lambda^l(\text{last}(r)) \) and the label of the root is \( \Lambda^l(\text{init}(1)) \). Observe that for every \( \sigma \in \Sigma_I \), the tree \( T_K \) can be obtained from \( T_{K,\sigma} \) by simply deleting the first component of the directions of \( T_{K,\sigma} \), and the second component of the labels of \( T_{K,\sigma} \).

Recall that the labeling of the root of a computation tree of \( K \) is not part of any trace (when it is not a sub-tree of another tree). Hence, in the definition below, we arbitrarily fix some letter \( \rho \in \Sigma_I \). Given a temporal logic formula \( \varphi \), over the atomic propositions \( AP \) where \( 2^{AP} = \Sigma_O \times \Sigma_I \), we have the following:

**Definition 1.** A hierarchical transducer \( K = \langle \Sigma_I, \Sigma_O, \langle K_1, \ldots, K_n \rangle \rangle \), with \( \text{Exit}_1 = 0 \), satisfies a formula \( \varphi \) (written \( K \models \varphi \)), iff the tree \( T_{K,\rho} \) satisfies \( \varphi \).

Observe that given \( \varphi \), finding a flat transducer \( K \) such that \( K \models \varphi \) is the classic synthesis problem studied (for LTL formulas) in [23].

A **library** \( \mathcal{L} \) is a finite set of hierarchical transducers with the same input and output alphabets. Formally, \( \mathcal{L} = \{ K^1, \ldots, K^\lambda \} \), and for every \( 1 \leq i \leq \lambda \), we have that \( K^i = \langle \Sigma_I, \Sigma_O, \langle K^i_1, \ldots, K^i_n \rangle \rangle \). Note that a transducer in the library can be a sub-transducer of another one, or share common sub-transducers with it. The set of transducers in \( \mathcal{L} \) that have no top-level exits is denoted by \( \mathcal{L}^0 = \{ K^i \in \mathcal{L} : \text{Exit}_{K^i}^i = 0 \} \), and its complement is \( \mathcal{L}^{\neq 0} = \mathcal{L} \setminus \mathcal{L}^0 \).

### 4 Hierarchical Synthesis

In this section, we describe our synthesis algorithm. We start by providing the algorithm with an initial library \( \mathcal{L}_0 \) of hierarchical transducers. A good starting point is to include in \( \mathcal{L}_0 \) all the atomic transducers, as well as any other relevant hierarchical transducers, for example from a standard library. We then proceed by synthesizing in rounds. At each round \( i \geq 0 \), the system designer provides a specification formula \( \varphi_i \), of the currently desired hierarchical transducer \( K^i \), which is then automatically synthesized using the transducers in \( \mathcal{L}_{i-1} \) as possible sub-transducers. Once a new transducer is synthesized it is added to the library, for use in subsequent rounds. Technically, the hierarchical transducer synthesized in the last round is the output of the algorithm.

**Input:** An initial library \( \mathcal{L}_0 \), and a list of specification formulas \( \varphi_1, \ldots, \varphi_m \)

**Output:** A hierarchical transducer satisfying \( \varphi_m \)

for \( i = 1 \) to \( m \) do

\[ L_i \leftarrow L_{i-1} \cup \{ K^i \} \]

end

return \( K^m \)

**Algorithm 1:** Hierarchical Synthesis Algorithm

The main challenge in implementing the above hierarchical synthesis algorithm is of course coming up with an algorithm for performing the synthesis step of a single round. As noted in Section 1, a transducer that was synthesized in a previous round has no top-level exits, which severely limits its ability to serve as a sub-transducer of another transducer. Our single-round algorithm must therefore address the problem of
synthesizing exits for such transducers. In Section 4.1, we give our core algorithm for single-round synthesis of a hierarchical transducer from a given library of hierarchical transducers. In Section 4.2, we address the problem of enforcing modularity, and add some more information regarding the synthesis of exits. Finally, in Section 4.3, we address the problem of synthesis with imperfect information.

4.1 Hierarchical Synthesis from a Library

We now formally present the problem of hierarchical synthesis from a library (that may have transducers without top-level exits) of a single temporal logic formula. Given a transducer \( \mathcal{K} = \langle \Sigma, \Sigma_0, (\mathcal{K}_1, \ldots, \mathcal{K}_n) \rangle \in \mathcal{L}^=\emptyset \), where \( \mathcal{K}_i = (W_i, \delta_i, \tau_i, \delta_i, \Lambda_i) \), and a set \( E \subseteq W_1 \), the transducer \( \mathcal{K}_E \) is obtained from \( \mathcal{K} \) by setting \( E \) to be the set of top-level exits, and removing all the outgoing edges from states in \( E \). Formally, \( \mathcal{K}_E = \langle \Sigma, \Sigma_0, (\langle W_1, \delta_1, \tau_1, \delta_1, \Lambda_1 \rangle, \mathcal{K}_2, \ldots, \mathcal{K}_n) \rangle \), where the transition relation \( \delta_1 \) is the restriction of \( \delta \) to sources in \( W_1 \setminus E \). For convenience, given a transducer \( \mathcal{K} \in \mathcal{L}^=\emptyset \), we sometimes refer to it as \( \mathcal{K}_{\text{Ex}} \). For every \( \mathcal{K} \in \mathcal{L} \), we assume some fixed ordering on the top-level states of \( \mathcal{K} \), and given a set \( E \subseteq W_1 \), and a state \( e \in E \), we denote by \( \text{id}(e, E) \) the relative position of \( e \) in \( E \), according to this ordering. Given a library \( \mathcal{L} \), and an upper bound \( el \in \mathbb{N} \) on the number of allowed top-level exits, we let \( \mathcal{L}^{el} = \mathcal{L}^{=\emptyset} \cup \{ \mathcal{K}_E : \mathcal{K} \in \mathcal{L}^{=\emptyset} \land |E| \leq el \} \). The higher the number \( el \), the more exits the synthesis algorithm is allowed to synthesize, and the longer it may take to run. As we show later, \( el \) should be at most polynomial\(^4\) in the size of \( \varphi \). In general, we assume that \( el \) is never smaller than the number of exits in any sub-transducer of any hierarchical transducer in \( \mathcal{L} \). Hence, for every \( \mathcal{K}_E \in \mathcal{L}^{el} \) and every \( e \in E \), we have that \( 1 \leq \text{id}(e, E) \leq el \).

Definition 2. Given a library \( \mathcal{L} \) and a bound \( el \in \mathbb{N} \), we say that:

- A hierarchical transducer \( \mathcal{K} = \langle \Sigma, \Sigma_0, (\mathcal{K}_1, \ldots, \mathcal{K}_n) \rangle \) is \( (\mathcal{L}, el) \)-composed if (i) for every \( 2 \leq i \leq n \), we have that \( \mathcal{K}_i \in \mathcal{L}^{el} \); (ii) if \( w \in W_1 \) is a top-level state, then the atomic transducer \( \mathcal{K}_{\Lambda_i(w)} \) is in \( \mathcal{L} \).
- A formula \( \varphi \) is \( (\mathcal{L}, el) \)-realizable if there is an \( (\mathcal{L}, el) \)-composed hierarchical transducer \( \mathcal{K} \) that satisfies \( \varphi \). The \( (\mathcal{L}, el) \)-synthesis problem is to find such a \( \mathcal{K} \).

Intuitively, an \( (\mathcal{L}, el) \)-composed hierarchical transducer \( \mathcal{K} \) is built by synthesizing its top-level sub-transducer \( \mathcal{K}_1 \), which specifies how to connect boxes that refer to transducers from \( \mathcal{L}^{el} \). To eliminate an unnecessary level of indirection, boxes that refer to atomic transducers are replaced by regular states. Note that this also solves the technical problem that, by definition, the initial state \( in_1 \) cannot be a box. This is also the reason why we assume from now on that every library has at least one atomic transducer. Note that for each transducer \( \mathcal{K}' \in \mathcal{L}^{=\emptyset} \) we can have as many as \( \Omega(|\mathcal{K}'|)^{el} \) copies of \( \mathcal{K}' \) in \( \mathcal{L}^{el} \), each with a different set of exit states. In Section 4.2 we show how, when we synthesize \( \mathcal{K} \), we can limit the number of such copies that \( \mathcal{K} \) uses to any desired value (usually one per \( \mathcal{K}' \)).

\(^4\) In practical terms, the exits of a sub-module represent its set of possible return values. Since finite state modules are usually not expected to have return values over large domains (such as the set of integers), we believe that our polynomial bound for \( el \) is not too restrictive.
**Connectivity trees** In the classical automata-theoretic approach to synthesis [23], synthesis is reduced to finding a regular tree that is a witness to the non-emptiness of a suitable tree automaton. Here, we also reduce synthesis to the non-emptiness problem of a tree automaton. However, unlike the classical approach, we build an automaton whose input is not a computation tree, but rather a system description in the form of a connectivity tree (inspired by the “control-flow” trees of [18]), which describes how to connect library components in a way that satisfies the specification formula. Specifically, given a library \( L = \{ K_1, ..., K_\lambda \} \) and a bound \( el \in \mathbb{N} \), connectivity trees represent hierarchical transducers that are \( \langle L, el \rangle \)-composed, in the sense that every regular \( \langle L, el \rangle \)-composed hierarchical transducer induces a connectivity tree, and vice versa. Formally, a connectivity tree \( T = (T, V) \) for \( L \) and \( el \), is an \( L^{el} \)-labeled complete \( \langle \{1, ..., el\} \times \Sigma \rangle \)-tree, where the root is labeled by an atomic transducer.

Intuitively, a node \( x \) with \( V(x) = K^E \) represents a top-level state \( q \) if \( K^E \) is an atomic transducer, and otherwise it represents a top-level box \( b \) that refers to \( K^E \). The label of a son \( x \cdot (ids(e,E), \sigma) \) specifies the destination of the transition from the exit \( e \) of \( b \) (or from a state \( q \), if \( K^E \) is atomic — in which case it has a single exit) when reading \( \sigma \). Sons \( x \cdot (i, e) \), for which \( i > |E| \), are ignored. Thus, a path \( \pi = (i_0, \sigma_{i_0}) \cdot (i_1, \sigma_{i_1}) \cdots \) in a connectivity tree \( T \) is called meaningful, iff for every \( j > 0 \), we have that \( i_j \) is not larger than the number of top-level exits of \( V(i_{j-1}, \sigma_{j-1}) \).

A connectivity tree \( T = (T, V) \) is regular if there is a flat transducer \( M = \langle \{1, ..., el\}, \Sigma, \{M, m_0, 0, \delta^T, \Lambda^T \rangle \rangle \), such that \( T \) is equal to the (lean) computation tree \( T_M \). A regular connectivity tree induces an \( \langle L, el \rangle \)-composed hierarchical transducer \( K \), whose top-level sub-transducer \( K^{el}_1 \) is basically a replica of \( M \) (see the full version at the authors’ web page for the reverse transformation). Every node \( m \in M \) becomes a state of \( K \) if \( \Lambda^T(m) \) is an atomic-transducer and, otherwise, it becomes a box of \( K \) which refers to the top-level sub-transducer of \( \Lambda^T(m) \). The destination of a transition from an exit \( e \) of a box \( m \), with \( \Lambda^T(m) = K^E \), when reading a letter \( \sigma \in \Sigma_j \), is given by \( \delta^T(m,(ids(e,E),\sigma)) \). If \( m \) is a state, then \( \Lambda^T(m) \) is an atomic transducer with a single exit and thus, the destination of a transition from \( m \) when reading a letter \( \sigma \in \Sigma_i \), is given by \( \delta^T(m,(1,\sigma)) \). For a box \( b \) of \( K_1 \), let \( \Lambda^T(b) = \langle \Sigma_i, \Sigma_O, \langle K_{b(1)}, ..., K_{b(n_b)} \rangle \rangle \), and denote by \( \text{sub}(b) = \{K_{b(1)}, ..., K_{b(n_b)}\} \) the set of sub-transducers of \( \Lambda^T(b) \), and by \( E(b) \) the set of top-level exits of \( \Lambda^T(b) \). Formally, \( K = \langle \Sigma_i, \Sigma_O, \langle K_1, ..., K_\lambda \rangle \rangle \), where \( K_1 = \langle W_1, B_1, m_0, \tau_1, \delta_1, \Lambda_1 \rangle \), and:

- \( W_1 = \{ w \in M : \exists \varsigma \in \Sigma_O \text{ s.t. } \Lambda^T(w) = K^{el}_\varsigma \} \). Note that since the root of a connectivity tree is labeled by an atomic transducer then \( m_0 \in W_1 \).
- \( B_1 = M \setminus W_1 \).
- The sub-transducers \( \{ K_2, ..., K_\lambda \} = \bigcup\{b \in B_1\} \cdot \text{sub}(b) \).
- For \( b \in B_1 \), we have that \( \tau_1(b) = i \), where \( i \) is such that \( K_i = K_{b(1)} \).
- For \( w \in W_1 \), and \( \sigma \in \Sigma_j \), we have that \( \delta_1(w,\sigma) = \delta^T(w,(1,\sigma)) \).
- For \( w \in W_1 \), and \( \sigma \in \Sigma_j \), we have that \( \delta_1((b,e),\sigma) = \delta^T(b,(ids(e,E(b)),\sigma)) \), for every \( e \in E(b) \) and \( \sigma \in \Sigma_j \).
- Finally, for \( w \in W_1 \) we have that \( \Lambda_1(w) = \varsigma \), where \( \varsigma \) is such that \( \Lambda^T(w) = K^{el}_\varsigma \).

**From synthesis to automata emptiness** Given a library \( L = \{ K^1, ..., K^\lambda \} \), a bound \( el \in \mathbb{N} \), and a temporal logic formula \( \varphi \), our aim is to build an APT \( A^T_\varphi \) such that \( A^T_\varphi \)
accepts a regular connectivity tree $T = \langle T, V \rangle$ iff it induces a hierarchical transducer $K$ such that $K \models \phi$. Recall that by Definition 1 and Theorem 1, $K \models \phi$ iff $T_{K, \rho}$ is accepted by the SAPT $A_{\phi}$. The basic idea is thus to have $A_{\phi}^T$ simulate all possible runs of $A_{\phi}$ on $T_{K, \rho}$. Unfortunately, since $A_{\phi}^T$ has as its input not $T_{K, \rho}$, but the connectivity tree $T$, this is not a trivial task. In order to see how we can solve this problem, we first have to make the following observation.

Let $T = \langle T, V \rangle$ be a regular connectivity tree, and let $K$ be the hierarchical transducer that it induces. Consider a node $u \in T_{K, \rho}$ with $\text{last}(u) = \langle (b, \text{in}_T(b)), \sigma \rangle$, where $b$ is some top-level box, or state, of $K$ that refers to some transducer $K^E$, (note that the root of $T_{K, \rho}$ is such a node). Observe that the sub-tree $T''$, rooted at $u$, represents the traces of computations of $K$ that start from the initial state of $K^E$, in the context of the box $b$. The sub-tree $\text{prune}(T'')$, obtained by pruning every path in $T''$ at the first node $\hat{u}$, with $\text{last}(\hat{u}) = \langle (b, e), \sigma \rangle$ for some $e \in E$ and $\sigma \in \Sigma_I$, represents the portions of these traces that stay inside $K^E$. Note that $\text{prune}(T'')$ is essentially independent of the context $b$ in which $K^E$ appears, and is isomorphic to the tree $T_{K^E, \sigma}$ (the isomorphism being to simply drop the component $b$ from every letter in the name of every node in $\text{prune}(T'')$). Moreover, every son $v$ (in $T_{K, \rho}$), of such a leaf $\hat{u}$ of $\text{prune}(T'')$, is of the same form as $u$, i.e., $\text{last}(v) = \langle (b', \text{in}_T(b')), \sigma' \rangle$, where $b' = \delta_1((b, e), \sigma')$ is a top-level box (or state) of $K$. It follows that $T_{K, \rho}$ is isomorphic to a concatenation of sub-trees of the form $T_{K^E, \sigma}$, where the transition from a leaf of one such sub-tree to the root of another is specified by the transition relation $\delta_1$, and is thus given explicitly by the connectivity tree $T$.

The last observation is the key to how $A_{\phi}^T$ can simulate, while reading $T$, all the possible runs of $A_{\phi}$ on $T_{K, \rho}$. The general idea is as follows. Consider a node $u$ of $T_{K, \rho}$ such that $\text{prune}(T'')$ is isomorphic to $T_{K^E, \sigma}$. A copy of $A_{\phi}^T$ that reads a node $y$ of $T$ labeled by $K^E$ can easily simulate, without consuming any input, all the portions of the runs of any copy of $A_{\phi}$ that start by reading $u$ and remain inside $\text{prune}(T'')$. This simulation can be done by simply constructing $T_{K^E, \sigma}$ on the fly and running $A_{\phi}$ on it. For every simulated copy of $A_{\phi}$ that reaches a leaf $\hat{u}$ of $\text{prune}(T'')$, (recall that $\text{last}(\hat{u})$ is of the form $\langle (b, e), \sigma \rangle$), the automaton $A_{\phi}^T$ sends copies of itself to the sons of $y$ in the connectivity tree, in order to continue the simulation on the different sub-trees rooted at sons of $\hat{u}$ in $T_{K, \rho}$. The simulation of a copy of $A_{\phi}$ that proceeds to a son $v = \hat{u} \cdot \langle (b', \text{in}_T(b')), \sigma' \rangle$, where $b'$ is a top-level box (or state) of $K$, is handled by a copy of $A_{\phi}^T$ that is sent to the son $z = y \cdot (\text{id}(e, E), \sigma')$.

Our construction of $A_{\phi}^T$ implements the above idea, with one main modification. In order to obtain optimal complexity in successive rounds of Algorithm 1, it is important to keep the size of $A_{\phi}^T$ independent of the size of the transducers in the library. Unfortunately, simulating the runs of $A_{\phi}$ on $T_{K^E, \sigma}$ on the fly would require an embedding of $K^E$ inside $A_{\phi}^T$. Recall, however, that no input is consumed by $A_{\phi}^T$ while running such a simulation. Hence, we can perform these simulations offline instead, in the process of building the transition relation of $A_{\phi}^T$. Obviously, this requires a way of summarizing the possibly infinite number of runs of $A_{\phi}$ on $T_{K^E, \sigma}$, which we do by employing the concept of summary functions from [6].

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5 Here we think of top-level states of $K$ as boxes that refer to atomic transducers.
First, we define an ordering ⪰ on colors by letting \( c \succeq c' \) when \( c \) is better, from the point of view of acceptance by \( A_\phi \), if \( c' \) is even then \( c \) is even and \( c \succeq c' \); and if \( c' \) is odd then either \( c \) is even, or \( c \) is also odd and \( c \leq c' \). We denote by \( \min^\succeq \) the operation of taking the minimal color, according to \( \succeq \), of a finite set of colors. Let \( A_q = (\Sigma_0 \times \Sigma_I, Q_0, q_0^0, \delta_0, F_0) \), let \( A_q' \) be the automaton \( A_q \) using \( q \in Q \) as an initial state, and let \( C \) be the set of colors used in the acceptance condition \( F_q \). Consider a run \( \langle T_r, r \rangle \) of \( A_q' \) on \( T_{R_E, \sigma} \). Note that if \( z \in T_r \) is a leaf, then \( \text{last}(r(z)) = ((e, \sigma'), q) \), where \( q \in Q_0^{\succeq} \) (i.e., \( q \) is not an \( \varepsilon \)-state), and \( e \in E \). We define a function \( g_r : E \times \Sigma_I \times Q_0^{\succeq} \rightarrow C \cup \{1\} \), called the summary function of \( \langle T_r, r \rangle \), which summarizes this run. Given \( h \in E \times \Sigma_I \times Q_0^{\succeq} \), if there is no leaf \( z \in T_r \) such that \( \text{last}(r(z)) = h \), then \( g_r(h) = \varepsilon \); otherwise, \( g_r(h) = c \), where \( c \) is the maximal color encountered by the copy of \( A_q \) which made the last progress towards satisfying the acceptance condition, among all copies that reach a leaf \( z \in T_r \) with \( \text{last}(r(z)) = h \). Formally, let \( \text{paths}(r, h) \) be the set of all the paths in \( \langle T_r, r \rangle \) that end in a leaf \( z \in T_r \), with \( \text{last}(r(z)) = h \). Then, \( g_r(h) = \varepsilon \) if \( \text{paths}(r, h) = \emptyset \) and, otherwise, \( g_r(h) = \min^\succeq \{\max C(\pi) : \pi \in \text{paths}(r, h)\} \).

Let \( Sf(K^E, \sigma, q) \) be the set of summary functions of the runs of \( A_q' \) on \( T_{R_E, \sigma} \). If \( T_{R_E, \sigma} \) has no leaves, then \( Sf(K^E, \sigma, q) \) contains only the empty summary function \( \varepsilon \).

For \( g \in Sf(K^E, \sigma, q) \), let \( g^{\varepsilon} = \{h \in E \times \Sigma_I \times Q_0^{\succeq} : g(h) \neq \varepsilon\} \). Based on the ordering \( \succeq \) we defined for colors, we can define a partial order \( \succeq \) on \( Sf(K^E, \sigma, q) \), by letting \( g \succeq g' \) if for every \( h \in (E \times \Sigma_I \times Q_0^{\succeq}) \) the following holds: \( g(h) = \varepsilon \) or \( g(h) \neq \varepsilon \neq g'(h) \) and \( g(h) \succeq g'(h) \). Observe that if \( r \) and \( r' \) are two non-rejecting runs, and \( g_r \succeq g_{r'} \), then extending \( r \) to an accepting run on a tree that extends \( T_{R_E, \sigma} \) is always no harder than extending \( r' \) - either because \( A_q \) has less copies at the leaves of \( r \), or because these copies encountered better maximal colors. Given a summary function \( g \), we say that a run \( \langle T_r, r \rangle \) achieves \( g \) if \( g_r \succeq g \); we say that \( g \) is feasible if there is a run \( \langle T_r, r \rangle \) that achieves it; and we say that \( g \) is relevant if it can be achieved by a memoryless\(^6\) run that is not rejecting (i.e., by a run that has no infinite path that does not satisfy the acceptance condition of \( A_q \)). We denote by \( \text{Rel}(K^E, \sigma, q) \subseteq Sf(K^E, \sigma, q) \) the set of relevant summary functions.

We are now ready to give a formal definition of the automaton \( A_q^{\varepsilon} \). Given a library \( L = \{K^1, \ldots, K^k\} \), a bound \( \ell \in \mathbb{N} \), and a temporal-logic formula \( \varphi \), let \( A_\varphi = (\Sigma_0 \times \Sigma_I, Q_0, q_0^0, \delta_0, F_0) \), let \( C = \{C_{\min}, \ldots, C_{\max}\} \) be the colors in the acceptance condition of \( A_\varphi \), and for \( K^E \in L^\ell \), let \( \Lambda^E \) be the labeling function of the top-level sub-transducer of \( K^E \). The automaton \( A_q^{\varepsilon} = (L^\ell, (\{1, \ldots, \ell\} \times \Sigma_I), (\Sigma_I \times Q_0^{\succeq} \times C) \cup \{q_0\}, q_0, \delta, \alpha) \), has the following elements:

- For every \( K^E \in L^\ell \) we have that \( \delta(q_0, K^E) = \delta((\rho, q_0^0, C_{\min}), K^E) \) if \( K^E \) is an atomic transducer and, otherwise, \( \delta(q_0, K^E) = \text{false} \).
- For every \( (\sigma, q, c) \in \Sigma_I \times Q_0^{\succeq} \times C \) and every \( K^E \in L^\ell \), we have \( \delta((\sigma, q, c), K^E) = \bigvee_{g \in \text{Rel}(K^E, \sigma, q)} \Lambda_{(\sigma, q, c)}((\text{id}(e, E), \sigma'), (\sigma', \delta_0(q), (\Lambda^E(e), \delta)), g(e, \delta, q))) \), where \( \Lambda = \emptyset \) if \( q \in Q_0^c \), and \( \Lambda = \emptyset \) if \( q \in Q_0^c \).
- \( \alpha(q_0) = C_{\min} \) and \( \alpha((\sigma, q, c)) = c \), for every \( (\sigma, q, c) \in \Sigma_I \times Q_0^{\succeq} \times C \).

\(^6\) A run of an automaton \( A \) is memoryless if two copies of \( A \) that are in the same state, and read the same input node, behave in the same way on the rest of the input.
The construction above implies the following lemma:

**Lemma 1.** $A^T_\emptyset$ accepts a regular connectivity tree $T = \langle T, V \rangle$ iff $T$ induces a hierarchical transducer $K$, such that $T_{K,\emptyset}$ is accepted by $A_\emptyset$.

**Proof (sketch).** Intuitively, $A^T_\emptyset$ first checks that the root of its input tree $T$ is labeled by an atomic proposition (and is thus a connectivity tree), and then proceeds to simulate all the runs of $A_\emptyset$ on $T_{K,\emptyset}$. A copy of $A^T_\emptyset$ at a state $(\sigma, q, c)$, that reads a node $y$ of $T$ labeled by $K^E$, considers all the non-rejecting runs of $A^T_\emptyset$ on $T_{K,\sigma}$, by looking at the set $\text{Rel}(K^E, \sigma, q)$ of summary functions for these runs. It then sends copies of $A^T_\emptyset$ to the sons of $y$ to continue the simulation of copies of $A_\emptyset$ that reach the leaves of $T_{K,\sigma}$.

The logic behind the definition of $\delta((\sigma, q, c), K^E)$ is as follows. Since every summary function $g \in \text{Rel}(K^E, \sigma, q)$ summarizes at least one non-rejecting run, and it is enough that one such run can be extended to a accepting run of $A_\emptyset$ on the remainder of $T_{K,\emptyset}$, we have a disjunction on all $g \in \text{Rel}(K^E, \sigma, q)$. Every $(e, \hat{\sigma}, \hat{q}) \in g^{\hat{\sigma}}$ represents one or more copies of $A_\emptyset$ at state $\hat{q}$ that are reading a leaf $\hat{u}$ of $T_{K,\sigma}$ with $\text{last}(\hat{u}) = (e, \hat{\sigma})$, and all these copies must accept their remainders of $T_{K,\emptyset}$. Hence, we have a conjunction over all $(e, \hat{\sigma}, \hat{q}) \in g^{\hat{\sigma}}$.

A copy of $A_\emptyset$ that starts at the root of $T_{K,\sigma}$ may give rise to many copies that reach a leaf $\hat{u}$ of $T_{K,\sigma}$ with $\text{last}(\hat{u}) = (e, \hat{\sigma})$, but we only need to consider the copy which made the least progress towards satisfying the acceptance condition, as captured by $g(e, \hat{\sigma}, \hat{q})$. To continue the simulation of such a copy on its remainder of $T_{K,\emptyset}$, we send to a son $y' = (\text{idx}(e, E), \sigma')$ of $y$ in the connectivity tree, whose label specifies where $K$ should go to from the exit $e$ when reading $\sigma'$, a copy of $A^T_\emptyset$ as follows. Recall that the leaf $\hat{u}$ corresponds to a node $u$ of $T_{K,\emptyset}$ such that $\text{last}(u) = ((b, e), \hat{\sigma})$ and $b$ is a top-level box of $K$ that refers to $K^E$. Also recall that every node in $T_{K,\emptyset}$ has one son for every letter $\sigma' \in \Sigma_l$. Hence, a copy of $A_\emptyset$ that is at state $\hat{q}$ and is reading $u$, sends one copy in state $q' = \delta_{\emptyset}(\hat{q}, (\Lambda^E(e), \hat{\sigma}))$ to each son of $u$, if $\hat{q} \in Q^e_\emptyset$; and only one such copy, to one of the sons of $u$, if $\hat{q} \in Q^e \emptyset$. This explains why $\bigoplus$ is a conjunction in the first case, and is a disjunction in the second. Finally, a copy of $A^T_\emptyset$ that is sent to direction $(\text{idx}(e, E), \sigma')$ carries with it the color $g(e, \hat{\sigma}, \hat{q})$, which is needed in order to define the acceptance condition. The color assigned to $q_0$ is of course arbitrary.

The core of the proof uses a game based approach. Recall that the game-based approach to model checking a flat system $S$ with respect to a branching-time temporal logic specification $\varphi$, reduces the model-checking problem to solving a game (called the membership game of $S$ and $A_\emptyset$) obtained by taking the product of $S$ with the alternating tree automaton $A_\emptyset$ [16]. In [6], this approach was extended to hierarchical structures, and it was shown there that given a hierarchical structure $S$ and an SAPT $A$, one can construct a hierarchical membership game $G_{S,A}$ such that Player 0 wins $G_{S,A}$ iff the tree obtained by unwinding $S$ is accepted by $A$. In particular, when $A$ accepts exactly all the tree models of a branching-time formula $\varphi$, the above holds iff $S$ satisfies $\varphi$. Furthermore, it is shown in [6] that one can simplify the hierarchical membership game $G_{S,A}$, by replacing boxes of the top-level arena with gadgets that are built using Player 0 summary functions, and obtain an equivalent flat game $G^f_{S,A}$.

Given a regular connectivity tree $T = \langle T, V \rangle$, that induces a hierarchical system $K$, we prove Lemma 1 by showing that the flat membership game $G^f_{S,A_\emptyset}$, where $S$ is a
hierarchical structure whose unwinding is the computation tree \( T_{X, \rho} \), is equivalent to the flat membership game \( G_{K', A'} \), of \( A_{\phi}^T \) and a Kripke structure \( K' \), whose unwinding is \( T \). Thus, \( A_{\phi} \) accepts \( T_{X, \rho} \) iff \( A_{\phi}^T \) accepts \( T \). The equivalence of these two games follows from the fact that they have isomorphic arenas and winning conditions. Consequently, our proof of Lemma 1 is mainly syntactic in nature, and amounts to little more than constructing the structures \( S \) and \( K' \), constructing the game \( G_{S, A_{\phi}} \), simplifying it to get \( G_{S, A'_{\phi}} \), and constructing the membership game \( G_{K', A'} \). The remaining technical details can be found in the full version on the authors’ web page. 

We now state our main theorem.

**Theorem 2.** The \( (\mathcal{L}, \text{el}) \)-synthesis problem is \( \text{EXPTIME}-\text{complete} \) for a \( \mu \)-calculus formula \( \varphi \), and is \( 2\text{EXPTIME}-\text{complete} \) for an \( \text{LTL} \) formula (for \( \text{el} \) that is at most polynomial in \(|\varphi|\) for \( \mu \)-calculus, or at most exponential in \(|\varphi|\) for \( \text{LTL} \)).

**Proof.** The lower bounds follow from the same bounds for the classical synthesis problem of flat systems [15, 25], and the fact that it is immediately reducible to our problem if \( \mathcal{L} \) contains all the atomic transducers. For the upper bounds, since an APT accepts some tree iff it accepts some regular tree (and \( A_{\phi}^T \) obviously only accepts trees which are connectivity trees), by Lemma 1 and Theorem 1, we get that an \( \text{LTL} \) or a \( \mu \)-calculus formula \( \varphi \) is \( (\mathcal{L}, \text{el}) \)-realizable iff \( L(A_{\phi}^T) \neq \emptyset \). Checking the emptiness of \( A_{\phi}^T \) can be done either directly, or by first translating it to an equivalent NPT \( A'_{\phi} \). For reasons that will become apparent in subsection 4.2, we choose the latter. Note that the known algorithms for checking the emptiness of an NPT are such that if \( L(A_{\phi}^T) \neq \emptyset \), then one can extract a regular tree in \( L(A_{\phi}^T) \) from the emptiness checking algorithm [24]. The upper bounds follow from the analysis given below of the time required to construct \( A_{\phi}^T \) and check for its non-emptiness.

By Theorem 1, the number of states \( |Q_\phi| \) and the index \( k \) of \( A_{\phi} \) is \( |Q_\phi| = 2^{|\phi|}, k = 2 \) for \( \text{LTL} \), and \( |Q_\phi| = O(|\varphi|), k = O(|\varphi|) \) for \( \mu \)-calculus. The most time consuming part in the construction of \( A_{\phi}^T \) is calculating for every \( (\mathcal{K}^E, \sigma, q) \in (\mathcal{L}^\text{el} \times \Sigma \times Q_\phi) \), the set \( \text{Rel}(\mathcal{K}^E, \sigma, q) \). Calculating \( \text{Rel}(\mathcal{K}^E, \sigma, q) \) can be done by checking for every summary function \( g \in \text{Sf}(\mathcal{K}^E, \sigma, q) \) if it is relevant. Our proof of Lemma 1 also yields that, by [6], the latter can be done in time \( O(|K| \cdot |Q_\phi|)^k \cdot (k + 1)^{|E| \cdot |Q_\phi|} \cdot k) \). Observe that the set \( \text{Sf}(\mathcal{K}^E, \sigma, q) \) is of size \( (k + 1)^{|E|} \), and that the number of transducers in \( \mathcal{L}^\text{el} \) is \( O(\lambda \cdot m^d) \), where \( m \) is the maximal size of any \( \mathcal{K} \in \mathcal{L} \). It follows that for an \( \text{LTL} \) (resp. \( \mu \)-calculus) formula \( \varphi \), the automaton \( A_{\phi}^T \) can be built in time at most polynomial in the size of the library, exponential in \( \text{el} \), and double exponential (resp. exponential) in \(|\varphi|\).

We now analyze the time it takes to check for the non-emptiness of \( A_{\phi}^T \). Recall that for every \( \eta \in (\mathcal{L}^\text{el} \times \Sigma_j \times Q_\phi) \), the set \( \text{Rel}(\eta) \) is of size at most \( (k + 1)^d \), and thus, the size of the transition relation of \( A_{\phi}^T \) is polynomial in \(|\mathcal{L}| \) and \(|\varphi|\), and exponential in \( \text{el} \). Checking the emptiness of \( A_{\phi}^T \) is done by first translating it to an equivalent NPT \( A'_{\phi} \).

By [20], given an APT with \(|Q|\) states and index \( k \), running on \( \Sigma \)-labeled \( \mathcal{D}^* \)-trees, one can build (in time polynomial in the descriptions of its input and output automata) an equivalent NPT with \((|Q| \cdot k)^{|\mathcal{D}|} \cdot |Q_\phi|\) states, an index \( O(|Q| \cdot k) \), and a transition relation of size \(|\Sigma| \cdot (|Q| \cdot k)^{|\mathcal{D}|} \cdot |Q_\phi|\). It is worth noting that this blow-up in the size of the
automaton is independent from the size of the transition relation of $\mathcal{A}_\phi^T$. By [16, 28], the emptiness of $\mathcal{A}_\phi'^T$ can be checked in time $|\Sigma| \cdot (|Q| \cdot k)^{O(|D| \cdot |Q|^2 \cdot k^2)}$ (and if it is not empty, a witness is returned). Recall that $|\Sigma| = |\mathcal{L}| = O(\lambda \cdot m e^t)$, and that $|D| = e l \cdot |\Sigma|$. By substituting the values calculated above for $|Q|$ and $k$, the theorem follows. $\square$

Note that in Algorithm 1, it is conceivable that the transducer $\mathcal{K}_i$ synthesized at iteration $i$ will be exponential (or even double-exponential for LTL) in the size of the specification formula $\phi_i$. At this point it is probably best to stop the process, refine the specifications, and try again. However, it is important to note that even if the process is continued, and $\mathcal{K}_i$ is added to the library, the time complexity of the succeeding iterations does not deteriorate since the single-round $\langle L, el \rangle$-synthesis algorithm is only polynomial in the maximal size $m$ of any transducer in the library.

### 4.2 Enforcing Modularity

In this section, we address two main issues that may hinder the efforts of our single-round $\langle L, el \rangle$-synthesis algorithm to synthesize a succinct hierarchical transducer $\mathcal{K}_i$. The first issue is that of ensuring that, when possible, $\mathcal{K}_i$ indeed makes use of the more complex transducers in the library (especially transducers synthesized in previous rounds) and does not rely too heavily on the less complex, or atomic, transducers. An obvious and most effective solution to this problem is to simply not have some (or all) of the atomic transducers present in the library. The second issue is making sure that $\mathcal{K}_i$ does not have too many sub-transducers, which can happen if it uses too many copies of the same transducer $\mathcal{K}'_i \in L^\mathcal{A}$, each with a different set of exits. We also discuss some other points of interest regarding the synthesis of exits. We address the above issues by constructing, for each constraint we want to enforce on the synthesized transducer $\mathcal{K}_i$, an APT $\mathcal{A}$, called the constraint monitor, such that $\mathcal{A}$ accepts only connectivity trees that satisfy the constraint. We then synthesize $\mathcal{K}_i$ by checking the non-emptiness not of $\mathcal{A}_\phi^T$, but of the product of $\mathcal{A}_\phi^T$ with all the constraints monitors. Note that a nondeterministic monitor (i.e., an NPT) of exponential size can also be used, without adversely affecting the time-complexity, if the product with it is taken after we translate the product of $\mathcal{A}_\phi^T$ and the other (polynomial) APT monitors, to an equivalent NPT.

A simple and effective way to enforce modularity in Algorithm 1 is that once a transducer $\mathcal{K}_i'$ is synthesized in round $i$, one incorporates in subsequent rounds a monitor that rejects any connectivity tree containing a node labeled by some key sub-transducers of $\mathcal{K}_i$. This effectively enforces any transducer synthesized using a formula that refers to atomic propositions present only in $\mathcal{K}_i'$ (and its disallowed sub-transducers) to use $\mathcal{K}_i'$, and not try to build its functionality from scratch. As to other ways to enforce modularity, the question of whether one system is more modular than another, or how to construct a modular system, has received many, and often widely different, answers. Here we only discuss how certain simple modularity criteria can be easily implemented on top of our algorithm. For example, some people would argue that a function that has more than, say, 10 consecutive lines of code in which no other function is called, is not modular enough. A monitor that checks that in no path in a connectivity tree there are more than 10 consecutive nodes labeled with an atomic transducer, can easily enforce such a criterion. We can even divide the transducers in the library into groups, based on how “high level” they are, and enforce lower counts on lower level groups. Essentially, every modularity criterion that can be checked by a polynomial APT, or an exponential
NPT, can be used. Enforcing one context-free property can also be done, albeit with an increase in the time complexity. Other non-regular criteria may be enforced by directly modifying the non-emptiness checking algorithm instead of by using a monitor, and we reserve this for future work.

As for the issue of synthesized exits, recall that for each transducer \( \mathcal{K}^e \in \mathcal{L}^e \), we can have as many as \( O(|\mathcal{K}^e|^2) \) copies of \( \mathcal{K}^e \) in \( \mathcal{L}^e \), each with a different set of exit states. Obviously, we would not like the synthesized transducer \( \mathcal{K} \) to use so many copies as sub-transducers. It is not hard to see that one can, for example, build an NPT of size \( O(|\mathcal{L}^e|) \) that guesses for every \( \mathcal{K} \in \mathcal{L}^e \) a single set of exits \( E \), and accepts a connectivity tree iff the labels of all the nodes in the tree agree with the guessed exits. Note that after the end of the current round of synthesis, we may choose to add \( \mathcal{K}^{E_i} \) to the library (in addition, or instead of \( \mathcal{K}^c \)).

Another point to note about the synthesis of exits is that while a transducer \( \mathcal{K} \) surely satisfies the formula \( \varphi_i \), it was synthesized for, \( \mathcal{K}^E \) may not. Consider for example a transducer \( \mathcal{K} \) which is simply a single state, labeled with \( p \), with a self loop. If we remove the loop and turn this state into an exit, it will no longer satisfy \( \varphi_i = p \land X p \) or \( \varphi_i = G p \). Now, depending on one’s point of view, this may be either an advantage (more flexibility) or a disadvantage (loss of original intent). We believe that this is mostly an advantage, however, in case it is considered a disadvantage, a few possible solutions come to mind. First, for example if \( \varphi_i = G p \), one may wish for \( \mathcal{K} \) to remain without exits and enforce \( E = \emptyset \). Another option, for example if \( \varphi_i = p \land X p \), is to synthesize in round \( i \) a modified formula like \( \varphi_i' = p \land \neg \text{exit} \land X (p \land \neg \text{exit}) \), with the thought of exits in mind. Yet another option is to add, at iterations after \( i \), a monitor that checks that if \( K^E \) is the label of a node in the connectivity tree then \( \varphi_i \) is satisfied. The monitor can check that \( \varphi_i \) is satisfied inside \( K^E \), in which case the monitor is a single state automaton, that only accepts if \( E \) is such that \( K^E \models \varphi_i \) (possibly using semantics over truncated paths [9]); alternatively, the monitor can check that \( \varphi_i \) is satisfied in the currently synthesized connectivity tree, starting from the node labeled by \( K^E \), in which case the monitor is based on \( \mathcal{K}^E_{0i} \).

### 4.3 Incomplete Information

A natural setting that was considered in the synthesis literature is that of incomplete information [15]. In this setting, in addition to the set of input signals \( I \) that the system can read, the environment also has internal signals \( H \) that the system cannot read, and one should synthesize a system whose behavior depends only on the readable signals, but satisfies a specification which refers also to the unreadable signals. Thus, the specification is given with respect to the alphabet \( \Sigma_I = 2^{I \cup H} \), but the behavior of the system must be the same when reading two letters that differ only in their \( H \) components. The main source of difficulty is that a finite automaton cannot decide whether or not a computation tree is of a system that behaves in a way which is consistent with its partial view of the input signals. However, since the automaton at the heart of our algorithm does not run on computation trees, but rather on connectivity trees, handling of incomplete information comes at no cost at all. All we have to do is to define the connectivity trees to be \( \mathcal{L}^c \)-labeled complete trees of \( [1,...,el] \times 2^{|H|} \)-trees, instead of \( ([1,...,el] \times 2^{|H|}) \)-trees to ensure that the synthesized transducer behaves in the same way on input letters that differ only in their hidden components (this of course implies that the expression \( \bigoplus_{\varphi_i \in \Sigma_I} \)
in the transition function of $A_T$ becomes $\bigoplus_{\sigma' \in 2^I} \sigma'$. Thus, our algorithm solves, with the same complexity, also the hierarchical synthesis problem with incomplete information.

5 Discussion

We presented an algorithm for the synthesis of hierarchical systems which takes as input a library of hierarchical transducers and a sequence of specification formulas. Each formula drives the synthesis of a new hierarchical transducer based on the current library, which contains all the transducers synthesized in previous iterations together with the starting library. The main challenge in this approach is to come up with a single-round synthesis algorithm that is able to efficiently synthesize the required transducer at each round. We have provided such an algorithm that works efficiently (i.e., not worst than the corresponding one for flat systems) and uniform (i.e., it can handle different temporal logic specifications, including the modal $\mu$-calculus). In order to ensure that the single-round algorithm makes real use of previously synthesized transducers we have suggested the use of auxiliary automata to enforce modularity criteria. We believe that by decoupling the process of enforcing modularity from the core algorithm for single-round synthesis we gain flexibility that allows one to apply different approaches to enforcing modularity, as well as future optimizations to the core synthesis algorithm.

References

A Modal Specification Theory
for Components with Data*

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Abstract. Modal specification is a well-known and widely used formalism used
as an abstraction theory for transition systems. Modal specifications are transi-
tion systems equipped with two types of transitions: must-transitions that are
mandatory to any implementation, and may-transitions that are optional. The
duality of transitions allows to develop a unique approach for both logical and
structural compositions, and eases the step-wise refinement process for building
implementations.

We propose Modal Specifications with Data (MSD), the first modal
specification theory with explicit representation of data. Our new theory includes all the essential
ingredients of a specification theory. As MSD are by nature potentially infinite-
state systems, we propose symbolic representations based on effective predicates.

Our theory serves as a new abstraction-based formalism for transition systems
with data.

1 Introduction

Modern IT systems are often large and consist of complex assemblies of numerous
reactive and interacting components. The components are often designed by independent
teams, working under a common agreement on what the interface of each component
should be. Consequently, the search for mathematical foundations which support compo-
sitional reasoning on interfaces is a major research goal. A framework should support
inferring properties of the global implementation, designing and advisedly reusing
components.

Interfaces are specifications and components that implement an interface are under-
stood as models/implementations. Specification theories should support various features
including (1) refinement, which allows to compare specifications as well as to replace a
specification by another one in a larger design, (2) structural composition, which allows
to combine specifications of different components, (3) logical conjunction, expressing
the intersection of the set of requirements expressed by two or more specifications, and

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last (4) a quotient operator that is dual to structural composition and allows synthesizing a component from a set of assumptions.

Among existing specification theories, one finds modal specifications [1], which are labeled transition systems equipped with two types of transitions: must-transitions that are mandatory for any implementation, and may-transitions which are optional for an implementation. Modal specifications are known to achieve a more flexible and easy-to-use compositional development methodology for CCS [2], which includes a considerable simplification of the step-wise refinement process proposed by Milner and Larsen. While being very close to logics (conjunction), the formalism takes advantage of a behavioral semantics allowing for easy composition with respect to process construction (structural composition) and synthesis (quotient). However, despite the many advantages, only a few implementations have been considered so far. One major problem is that contrary to other formalisms based on transition systems, there exists no theory of modal specification equipped with rich information such as data variables.

In this paper, we add a new stone to the cathedral of results on modal specifications [3, 4], that is we propose the first such theory equipped with rich data values. Our first contribution is to design a semantical version of modal specifications whose states are split into locations and valuations for possibly infinite-domain variables. For every component, we distinguish between local variables, that are locally controlled by the component, and uncontrolled variables that are controlled by other components and can be accessed, but not modified. Combining variables with sets of actions labeling transitions offers a powerful set of communication primitives that cannot be captured by most existing specification theories. We also propose a symbolic predicate-based representation of our formalism. We consider effective predicates that are closed under conjunction, union, and membership—classical assumptions in existing symbolic theories (e.g. [5]). While the semantic level is possibly infinite-state, the syntactical level permits us to reason on specifications just like one would with the original modal specifications, but with the additional power of rich data.

Continuing our quest, we study modal refinement between specifications. Refinement, which resembles simulation between transition systems, permits to compare sets of implementations in a syntactic manner. Modal refinement is defined at the semantic level, but can also be checked at the symbolic level. We propose a predicate abstraction approach that simplifies the practical complexity of the operation by reducing the number of states and simplifying the predicates. This approach is in line with the work of Godefroid et al. [6], but is applied to specification-based verification rather than to model checking.

We then propose definitions for both logical and structural composition, on the level of symbolic representations of specifications. These definitions are clearly not direct extensions of the ones defined on modal specifications as behaviors of both controlled and uncontrolled variables have to be taken into account. As usual, structural composition offers the property of independent implementability, hence allowing for elegant step-wise refinement. In logical composition, two specifications which disagree on their requirements can be reconciled by synthesizing a new component where conflicts have been removed. This can be done with a symbolic pruning of bad states, which terminates if the system is finite-state, or if the structure of the transition system induced by the
specification relies, for instance, on a well-quasi order [7]. Finally, we also propose a quotient operation, that is the dual operation of structural composition, which works for a subclass of systems, and we discuss its limitation. This operator, absent from most existing behavioral and logical specification theories, allows synthesizing a component from a set of assumptions.

In Sect. 2 we introduce modal specifications with data and their finite symbolic representations, refinement, an implementation relation and consistency. In Sect. 3 we define the essential operators of every specification theory, that is parallel composition, conjunction and quotient. For verification of refinement between infinite-state specifications we propose in Sect. 4 an approach based on predicate abstraction techniques. We summarize related works in Sect. 5 and conclude in Sect. 6.

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2 Modal Specifications with Data

We will first introduce specifications which are finite symbolic representations of modal specifications with data. We will then propose modal refinement and derive an implementation relation and a consistency notion.

In the following, $\mathcal{P}(M)$ denotes the powerset of $M$, $\mathcal{P}_{\geq 1}(M) = \mathcal{P}(M) \setminus \{\emptyset\}$, and the union of two disjoint sets is denoted by $M \uplus N$, which is $M \cup N$ with $M \cap N = \emptyset$.

Let $\forall$ be a fixed set of variables, each variable ranging over a fixed domain $\mathbb{D}$. For a given subset $V \subseteq \forall$, a data state $s$ over $V$ is a mapping $s : V \rightarrow \mathbb{D}$. If $V = \{x_1, x_2, \ldots, x_n\}$ and $d_1, d_2, \ldots, d_n \in \mathbb{D}$, we write $[x_1 \mapsto d_1, x_2 \mapsto d_2, \ldots, x_n \mapsto d_n]$ for the data state $s$ which maps every $x_i$ to $d_i$, for $1 \leq i \leq n$. We write $[V]$ for the set of all possible data states over $V$. For disjoint sets of variables $V_1$ and $V_2$ and data states $s_1 \in [V_1]$ and $s_2 \in [V_2]$, the operation $(s_1 \cdot s_2)$ composes the data states resulting in a new state $s = (s_1 \cdot s_2) \in [V_1 \uplus V_2]$, such that $s(x) = s_1(x)$ for all $x \in V_1$ and $s(x) = s_2(x)$ for all $x \in V_2$. This is naturally lifted to sets of states: if $S_1 \subseteq [V_1]$ and $S_2 \subseteq [V_2]$ then $(S_1 \cdot S_2) = \{(s_1 \cdot s_2) | s_1 \in S_1, s_2 \in S_2\} \subseteq [V_1 \uplus V_2]$.

Like in the work of de Alfaro et al. [8] we define specifications with respect to an assertion language allowing suitable predicate representation. Given a set $V$ of variables, we denote by $\text{Pred}(V)$ the set of first-order predicates with free variables in $V$; we assume that these predicates are written in some specified first-order language with existential ($\exists$) and universal ($\forall$) quantifiers and with interpreted function symbols and predicates; in our examples, the language contains the usual arithmetic operators and boolean connectives ($\lor, \land, \neg, \Rightarrow$). Syntactic equality of predicates is written with the symbol $\equiv$. Given a set of variables $V$ we denote by $(V)'$ an isomorphic set of ‘primed’ variables from $V$: so if $x \in V$ then $(x)' \in (V)'$. We use this construction to represent pre- and post-values of variables. A variable $(x)' \in (V)'$ represents the next state value of the variable $x \in V$. Given a formula $\varphi \in \text{Pred}(V)$ and a data state $s \in [V]$, we write $\varphi(s)$ if the predicate formula $\varphi$ is true when its free variables are interpreted as specified by $s$. Given a formula $\psi \in \text{Pred}(V_1 \uplus (V_2)')$ and states $s_1 \in [V_1]$, $s_2 \in [V_2]'$, we often write $\psi(s_1, s_2)$ for $\psi(s_1 \cdot t_2)$ where $t_2 \in [(V_2)']$ such that $t_2((x)') = s_2(x)$ for all $x \in V_2$. Given a predicate $\varphi \in \text{Pred}(V)$, we write $(\varphi)' \in \text{Pred}((V)'')$ for the
predicate obtained by substituting $x$ with $(x)'$ in $\varphi$, for all $x \in V$. We write $\llbracket \varphi \rrbracket$ for the set $\{s \in V \mid \varphi(s)\}$ which consists of all states satisfying $\varphi \in \text{Pred}(V)$ (for predicates with primed and unprimed variables), and $\varphi$ is consistent if $\llbracket \varphi \rrbracket \neq \emptyset$. We write $\exists V \varphi$ meaning existential quantification of $\varphi$ over all variables in the set $V$, and similar for universal quantification. Finally, for a predicate $\psi \in \text{Pred}(V \cup (V')')$, we write $\circ \psi$ for $\exists (V_2)' \psi$, and $\psi^o$ for $\forall i \psi$.

Our theory enriches modal automata with variables. Specifications not only express constraints on the allowed sequences of actions, but also their dependence and effect on the values of variables. Like in the loose approach of modal specifications [1] which allows under-specification using may and must modalities on transitions, we allow loose specification of the effects of actions on the data state. From a given location and a given data state, a transition to another location is allowed to lead to several next data states. Unlike in modal specifications, variables are observable in our framework, allowing for modeling shared variable communication.

A signature $\text{Sig} = (\Sigma, V^L, V^G)$ determines the alphabet of actions $\Sigma$ and the set of variables $V = V^L \cup V^G$ of an interface. The variables in $V^L$ are local (controlled) variables, owned by the interface and visible to any other component. $V^G$ contains the uncontrolled variables owned by the environment, which are read-only for the interface.

Specifications are finite modal transition systems where transitions are equipped with predicates. A transition predicate $\psi \in \text{Pred}(V \cup (V')')$ relates a previous state, determined by all controlled and uncontrolled data states, with the next possible controlled data state.

**Definition 1.** A specification is a tuple $A = (\text{Sig}, \text{Loc}, \ell^0, \varphi^0, E_\circ, E_\square)$ where $\text{Sig} = (\Sigma, V^L, V^G)$ is a signature, $\text{Loc}$ is a finite set of locations, $\ell^0 \in \text{Loc}$ is the initial location, $\varphi^0 \in \text{Pred}(V^L)$ is a predicate on the initial local state, and $E_\circ, E_\square$ are finite may- and must-transition relations respectively:

$$E_\circ, E_\square \subseteq \text{Loc} \times \Sigma \times \text{Pred}(V \cup (V')') \times \text{Loc}.$$  

Given a specification $A$, locations $\ell, \ell' \in \text{Loc}$, and action $a \in \Sigma$, we refer to the set of transition predicates on may-transitions by $\text{May}^a(\ell, \ell') = \{ \psi \mid (\ell, a, \psi, \ell') \in E_\circ \}$ and on must-transitions by $\text{Must}^a(\ell, \ell') = \{ \psi \mid (\ell, a, \psi, \ell') \in E_\square \}$.

**Example 1.** Consider a specification of a print server, shown in Fig. 1. Must-transitions are drawn with solid arrows and may-transitions with dashed ones. Every solid arrow representing a must-transition has an implicit may-transition shadowing it which is not shown. Every transition is equipped with a transition predicate over unprimed variables, referring to the pre-state, and primed variables, referring to the poststate. The print server receives new print jobs (newPrintJob), stores them and assigns them either a low or high priority; the numbers of low and high priority jobs are modeled by controlled variables $l$ and $h$, respectively; $l$ and $h$ are natural numbers. A job with low priority can also be reclassified to high priority (incPriority). The printer server can send (send) a job to a printer, and then wait for the acknowledgment (ack). In state $\ell_1$, if there is a job with high priority and the uncontrolled boolean variable priorityMode is true, then there must be a send transition. The specification is loose in the sense that if a second print job is received in state $\ell_1$, then the behavior is left unspecified.
We now define the kind of transition systems which will be used for formalizing the semantics of specifications. A specification is interpreted as a variant of modal transition systems where the state space is formed by the cartesian product $\text{Loc} \times \mathbb{[V^L]}$, i.e. a state is a pair $(\ell, s)$ where $\ell \in \text{Loc}$ is a location and $s \in \mathbb{[V^L]}$ is a valuation of the controlled variables. To motivate the choice of the transition relations in the semantics of specifications, we first describe the intended meaning of may- and must-transitions.

A may-transition $(\ell, a, \psi, \ell') \in E_\Diamond$ in the specification expresses that in any implementation, in any state $(\ell, s)$ and for any guard $g \in \mathbb{[V^G]}$ (that is a valuation of uncontrolled variables $V^G$) the implementation is allowed to have a transition with guard $g$ and action $a$ to a next state $(\ell', s')$ such that $\psi(s \cdot g, s')$. The interpretation of a must-transition $(\ell, a, \psi, \ell') \in E_\Box$ is a bit more involved: Any implementation, in state $(\ell, s)$, and for any guard $g \in \mathbb{[V^G]}$, if there is a valuation $s' \in \mathbb{[V^L]}$ such that $\psi(s \cdot g, s')$, then the implementation is required to have a transition from state $(\ell, s)$ with guard $g$ and action $a$ to at least some state $t'$ such that $\psi(s \cdot g, t')$. The requirement expressed by must-transitions cannot be formalized by standard modal transition systems, but fortunately, a generalization called disjunctive modal transition systems introduced in [9] can precisely capture these requirements. May-transitions target (as usual) only one state, but must-transitions branch to several possible next states (thus must-transitions are hypertransitions), with an existential interpretation: there must exist at least one transition with some target state which is an element from the set of target states of the hypertransition.

**Definition 2.** A modal specification with data (MSD) is a tuple

$$S = (\text{Sig}, \text{Loc}, \ell^0, S^0, \rightarrow_\Diamond, \rightarrow_\Box)$$

where Sig, Loc, $\ell^0$ are like in Def. 1, $S^0 \subseteq \mathbb{[V^L]}$ is a set of initial data states, and $\rightarrow_\Diamond, \rightarrow_\Box \subseteq \text{Loc} \times \mathbb{[V^L]} \times \mathbb{[V^G]} \times \Sigma \times (\text{Loc} \times \mathcal{P}_{\geq 1}(\mathbb{[V^L]}))$ are the may- ($\rightarrow_\Diamond$) and must- ($\rightarrow_\Box$) transition relations such that every may-transition targets a single state: if $(\ell, s, g, a, (\ell', s')) \in \rightarrow_\Diamond$ then $|S'| = 1$.

A state $(\ell, s) \in \text{Loc} \times \mathbb{[V^L]}$ is called syntactically consistent iff targets reachable by must-transitions are also reachable by may-transitions: if $(\ell, s, g, a, (\ell', S')) \in \rightarrow_\Box$ then $(\ell, s, g, a, (\ell', \{s'\})) \in \rightarrow_\Diamond$ for all $s' \in S'$. S is syntactically consistent iff all states are syntactically consistent, and the set of initial data states is nonempty, i.e. $S^0 \neq \emptyset$. 

![Fig. 1. Abstract specification $P$ of a print server.](image)
May-transitions \((\ell, s, g, a, (\ell', S')) \in \rightarrow_{\phi}\) are often written \((\ell, s) \xrightarrow{a}_{\phi} (\ell', S')\), and similarly for must-transitions.

We can now define formally how a specification translates to its semantics in terms of an MSD. As already described above, the semantics of a may-transition of the specification is given by the set of may-transitions pointing to single admissible target states, and a must-transition gives rise to (must-)hypertransitions targeting all the admissible poststates.

**Definition 3.** The semantics of a specification \(A = (\text{Sig}, \text{Loc}, \ell^0, \phi^0, E_\phi, E_{\square})\) is given by the MSD \(\langle A \rangle_{\text{sem}} = (\text{Sig}, \text{Loc}, \ell^0, S^0, \rightarrow_{\phi}, \rightarrow_{\square})\) where \(S^0 = \llbracket \phi^0 \rrbracket\) and the transition relations are defined as follows. For each \(\ell, \ell' \in \text{Loc}, s, s' \in \llbracket V^L \rrbracket, g \in \llbracket V^G \rrbracket\), and \(a \in \Sigma\):

1. If \((\ell, a, \psi, \ell') \in E_\phi\) and \(\psi(s \cdot g, s')\) then \((\ell, s) \xrightarrow{a}_{\phi} (\ell', \{s'\})\).
2. If \((\ell, a, \psi, \ell') \in E_{\square}\) and \(\psi(s \cdot g, s')\) then \((\ell, s) \xrightarrow{a}_{\square} (\ell', \{s' \in \llbracket V^L \rrbracket | \psi(s \cdot g, \ell')\})\).

A specification \(A\) is called **syntactically consistent** iff its semantics \(\langle A \rangle_{\text{sem}}\) is syntactically consistent. In the following we will always assume that specifications and MSD are syntactically consistent.

**Example 2.** An excerpt of the semantics of our abstract specification of the print server (see Fig. 1) can be seen Fig. 2. As before, we draw must-transitions with a solid arrow, and has an implicit set of may-transitions shadowing it which are not shown, i.e. for each target \((\ell, S')\) of a must-transition and each \(s \in S'\) there is a may-transition with the same source state and with target state \((\ell, \{s\})\).

The first must-transition \((\ell_0, \text{newPrintJob}, (l') + (h') = 1, \ell_1) \in E_{\square}\) of the print server specification gives rise to the transitions shown in Fig. 2. Any new print job must be stored in either \(l\) or \(h\) but which one is not yet fixed by the specification. Thus in the semantics this is expressed as a disjunctive must-transition to the unique location \(\ell_1\) and the next possible data states \([l \mapsto 1, h \mapsto 0]\) and \([l \mapsto 0, h \mapsto 1]\).

A refinement relation allows to relate a concrete specification with an abstract specification. Refinement should satisfy the following substitutability property: If \(A\) refines \(B\) then replacing \(B\) with \(A\) in a context \(C[\cdot]\) gives a specification \(C[A]\) refining \(C[B]\). Refinement will be a precongruence, i.e. it is compatible with the structural and logical operators on specifications in the above sense.

Our definition of refinement is based on modal refinement [10, 9] for (disjunctive) modal transition systems, where the may-transitions determine which actions are permitted in a refinement while the must-transitions specify which actions must be present in a refinement and hence in any implementation. We adapt it with respect to data states.
Example 3. We motivate our adoption of modal refinement to take into account data states with the help of a small example shown in Fig. 3. We draw may-transitions with a dashed arrow, and must-transitions with a solid arrow. Every must-transition has an implicit set of may-transitions shadowing it which are not shown. The MSD T (to the right) has two initial states, both having \( \ell_0 \) as the initial location. The must-transition starting from \((\ell_0, s_0)\) expresses that in any implementation there must be a transition leading to at least one of the states \((\ell_1, s_1)\) and \((\ell_1, s_2)\). The MSD T can be refined to the MSD S (by dropping one may-transition and turning one may-transition to a must-transition), and then S is refined by the MSD R, by refining the must-transition \((\ell_0', s_0, g_1, a, (\ell_1', \{s_1, s_2\}))\) in S to the must-transition \((\ell_0'', s_0, g_1, a, (\ell_1', \{s_1\}))\) in R, and by strengthening the transition with guard \(g_3\) and action \(c\) to a must-transition.

Definition 4. Let \( T_1 = (\text{Sig}, \text{Loc}_1, \ell_1^0, S_1^0, \rightarrow_{\circ, 1}, \rightarrow_{\Box, 1}) \) and \( T_2 = (\text{Sig}, \text{Loc}_2, \ell_2^0, S_2^0, \rightarrow_{\circ, 2}, \rightarrow_{\Box, 2}) \) be MSD over the same signature \( \text{Sig} = (\Sigma, V^L, V^G) \). A relation \( R \subseteq \text{Loc}_1 \times \text{Loc}_2 \times \llbracket V^L \rrbracket \) is a refinement relation iff for all \((\ell_1, \ell_2, s) \in R:\)

i. Whenever \((\ell_1, s) \xrightarrow{a}{\circ, 1} (\ell_1', \{s'\})\) then there exists \((\ell_2, s) \xrightarrow{a}{\circ, 2} (\ell_2', \{t'\})\) such that \(s' = t'\) and \((\ell_1, \ell_2', s') \in R\).

ii. Whenever \((\ell_2, s) \xrightarrow{a}{\Box, 2} (\ell_2', S_2')\) then there exists \((\ell_1, s) \xrightarrow{a}{\circ, 1} (\ell_1', S_1')\) such that \(S_1' \subseteq S_2'\) and \((\ell_1, \ell_2', s') \in R\) for all \(s' \in S_1'\).

We say that \( T_1 \) refines \( T_2 \), written \( T_1 \preceq_{\text{sem}} T_2 \), iff \( S_1^0 \subseteq S_2^0 \) and there exists a refinement relation \( R \) such that for any \( s \in S_1^0 \) also \((\ell_1', \ell_2, s) \in R\). A specification \( A_1 \) refines another specification \( A_2 \), written \( A_1 \preceq_{\text{sem}} A_2 \), iff \((A_1)_{\text{sem}} \preceq_{\text{sem}} (A_2)_{\text{sem}}\).

The refinement relation is a preorder on the class of all specifications. Refinement can be checked in polynomial time in the size of the state space of the MSD (for variables with finite domains). In general the domain may be infinite, or prohibitively large, so in Sect. 4 we revisit the question of refinement checking using abstraction techniques.

Example 4. The semantics of our abstract print server specification, shown in Fig. 2, can be refined as shown in Fig. 4. Now, both must-transitions point to the location \( \ell_1 \) with the data state \([l \mapsto 1, h \mapsto 0]\) which means that any new incoming print job is assigned a low priority, independent of the uncontrolled variable \(\text{priorityMode}\).

An MSD for which the conditions (1) \(\rightarrow_\circ = \rightarrow_\Box\) and (2) \(|S^0| = 1\) are satisfied, can be interpreted as (an abstraction of) an implementation: there are no design choices left open as (1) all may-transitions are covered by must-transitions and (2) there is only one initial data state possible. Any MSD for which the conditions (1) and (2) are satisfied, is called transition system with data (TSD) in the following. Note that TSD cannot be
strictly refined, i.e. for any TSD $I$ and any MSD $S$ with the same signature, $S \leq_{sem} I$ implies $I \leq_{sem} S$.

An implementation relation connects specifications to implementations (given as TSD) satisfying them. We can simply use refinement as the implementation relation. Given a specification $A$ and some TSD $I$, we write $I \models A$ for $I \leq_{sem} \langle A \rangle_{sem}$, so our implementation $I$ is seen as the model which satisfies the property expressed by the specification $A$. Now the set of implementations of a specification is the set of all its refining TSD: given a specification $A$, we define $\text{Impl}(A) = \{ I \mid I \models A \}$.

Our implementation relation $\models$ immediately leads to the classical notion of consistency as existence of models. A specification $A$ is consistent iff $\text{Impl}(A)$ is non-empty. Consequently, as modal refinement is reflexive, any specification $A$ for which $\langle A \rangle_{sem}$ is a TSD, is consistent.

By transitivity, modal refinement entails implementation set inclusion: for specifications $A$ and $B$, if $A \leq B$ then $\text{Impl}(A) \subseteq \text{Impl}(B)$. The relation $\text{Impl}(A) \subseteq \text{Impl}(B)$ is sometimes called thorough refinement [11]. Just like for modal transition systems, thorough refinement does not imply modal refinement in general [12]. To establish equivalence we follow [13] by imposing a restriction on $B$, namely that it is deterministic. An MSD is deterministic if

1. if $(\ell, s, g, a, (\ell', S')) \in \rightarrow_{\Box}$ then $(\ell', S') = (\ell'', S'')$,
2. if $(\ell, s, g, a, (\ell', S')) \in \rightarrow_{\Diamond} \cup \rightarrow_{\Box}$ then $\ell' = \ell''$.

A specification $B$ is deterministic, if the MSD $\langle B \rangle_{sem}$ is deterministic. Note that for may-transitions, determinism only requires that for the same source state, guard and action, the transition leads to a unique next location. The reason why this is sufficient is that modal refinement explicitly distinguishes states by their data state part: two states $(\ell, s)$ and $(\ell', s')$ can only be related if their data state parts $s$, $s'$ coincide.

Now, turning back to the relationship of modal refinement and inclusion of implementation sets (thorough refinement), we can prove the following theorem. Under the restriction of determinism of the refined (abstract) specification we can prove completeness of refinement. This theorem effectively means that modal refinement, as defined for MSD, is characterized by set inclusion of admitted implementations.

**Theorem 1.** Let $A$ and $B$ be two specifications with the same signature such that $B$ is deterministic. Then $A \leq B$ if and only if $\text{Impl}(A) \subseteq \text{Impl}(B)$.

### 3 Compositional Reasoning

In this section we propose all the essential operators on specifications a good specification theory should provide. We will distinguish between structural and logical composition.
Structural composition mimics the classical composition of transition systems at the specification level. Logical composition allows to compute the intersection of sets of models and hence can be used to represent the conjunction of requirements made on an implementation. Furthermore we will introduce a quotient operator which is the dual operator to structural composition.

From now on, we assume that for any two specifications with the signatures \( \text{Sig}_1 = (\Sigma_1, V_1^L, V_1^G) \) and \( \text{Sig}_2 = (\Sigma_2, V_2^L, V_2^G) \), respectively, we can assume that \( \Sigma_1 = \Sigma_2 \) and \( V_1^L \sqcup V_2^G = V_2^L \sqcup V_2^G \). This is not a limitation, as one can apply the constructions of [4] to equalize alphabets of actions and sets of variables.

**Parallel composition.** Two specifications \( A_1 \) and \( A_2 \) with \( \text{Sig}_1 = (\Sigma_1, V_1^L, V_1^G) \), \( \text{Sig}_2 = (\Sigma_2, V_2^L, V_2^G) \), respectively, are composable iff \( V_1^L \cap V_2^L = \emptyset \). Then their signatures can be composed in a straightforward manner to the signature

\[
\text{Sig}_1 \parallel \text{Sig}_2 = (\Sigma_1, V_1^L \cup V_2^L, (V_1^G \cup V_2^G) \setminus (V_1^L \cup V_2^L))
\]

in which the set of controlled variables is the union of the sets of controlled variables of \( A_1 \) and \( A_2 \), and the set of uncontrolled variables consists of all those uncontrolled variables of \( A_1 \) and \( A_2 \) which are controlled neither by \( A_1 \) nor by \( A_2 \).

**Definition 5.** Let \( A_1 \) and \( A_2 \) be two composable specifications. The parallel composition of \( A_1 \) and \( A_2 \) is defined as the specification

\[
A_1 \parallel A_2 = (\text{Sig}_1 \parallel \text{Sig}_2, \text{Loc}_1 \times \text{Loc}_2, (\ell_1^0, \ell_2^0), \varphi_1^0 \land \varphi_2^0, E_0, E_\Box)
\]

where the transition relations \( E_0 \) and \( E_\Box \) are the smallest relations satisfying the rules:

1. if \( (\ell_1, a, \psi_1, \ell_1') \in E_0 \) and \( (\ell_2, a, \psi_2, \ell_2') \in E_0 \) then
   \[
   (\ell_1, \ell_2), a, \psi_1 \land \psi_2, (\ell_1', \ell_2') \in E_0.
   \]
2. if \( (\ell_1, a, \psi_1, \ell_1') \in E_\Box \) and \( (\ell_2, a, \psi_2, \ell_2') \in E_\Box \) then
   \[
   (\ell_1, \ell_2), a, \psi_1 \land \psi_2, (\ell_1', \ell_2') \in E_\Box.
   \]

Composition of specifications, similar to the classical notion of modal composition for modal transition systems [10], synchronizes on matching shared actions and only yields a must-transition if there exist corresponding matching must-transitions in the original specifications. Composition is commutative (up to isomorphism) and associative. Our theory supports independent implementability of specifications, which is a crucial requirement for any compositional specification framework [14].

**Theorem 2.** Let \( A_1, A_2, B_1, B_2 \) be specifications such that \( A_1 \) and \( B_1 \) are composable. If \( A_1 \leq A_2 \) and \( B_1 \leq B_2 \), then \( A_1 \parallel B_1 \leq A_2 \parallel B_2 \).

The analog of parallel composition on the level of specifications is parallel composition \( ||_{\text{sem}} \) on the level of MSD which is a straightforward translation of the above symbolic rules. In fact one can prove that both parallel compositions \( || \) and \( ||_{\text{sem}} \) are equivalent, i.e. that \( (A_1 || A_2)_{\text{sem}} = (A_1)_{\text{sem}} ||_{\text{sem}} (A_2)_{\text{sem}} \) for any two composable specifications \( A_1, A_2 \).
Remark 1. Interface theories based on transition systems labeled with input/output actions usually involve a notion of compatibility, which is a relation between interfaces determining whether two components can work properly together. Since the present theory does not have a notion of input/output it is enough to require that two components are composable, i.e. that their local variables do not overlap. A pessimistic input/output compatibility notion has been proposed in our previous work [15]. Optimistic input/output compatibility based on a game semantics allows computing all the environments in which two components can work together. Following our recent works in [16, 4], one can enrich labels of transitions in the present theory with input and output and apply the same game-based semantics in order to achieve an optimistic composition.

Syntactical consistency. Our next two specification operators, conjunction and quotient, may yield specifications which are syntactically inconsistent, i.e. either there is no legal initial data state or there are states with a must-transition but without corresponding may-transition.

In general, given a specification $A$, syntactic consistency implies consistency, i.e. $\text{Impl}(A) \neq \emptyset$, but in general, the reverse does not hold. However, every consistent specification can be “pruned” to a syntactically consistent one, by pruning backwards from all syntactically inconsistent states, removing states which have to reach some of the “bad” states. Pruning will be shown to preserve the set of implementations.

For a specification $A = (\text{Sig}, \text{Loc}, \ell^0, \varphi^0, E_0, E_{[\_]} )$, the pruning (or reduction) of $A$, denoted by $p(A)$, is done as follows. Let $B : \text{Loc} \rightarrow \text{Pred}(V^L)$ be a mapping of locations to predicates over the local variables. We define a predecessor operation, iteratively computing all states that are forced to reach a “bad” state. Define a weakest precondition predicate, for $\psi \in \text{Pred}(V \cup (V^L)^\prime)$, $\varphi \in \text{Pred}(V^L)$, by

\[
wp_{\psi}[\varphi] \equiv \exists V^G \cdot \psi \land (\forall(V^L)^\prime, \psi \Rightarrow (\varphi)^\prime)
\]

which computes the largest set of local states such that there exists an uncontrolled state $g \in [V^G]$ such that $\psi$ maps to at least one next state, and all next states satisfy $\varphi$. Then

\[
\text{predec}(B)(\ell) \equiv B(\ell) \lor \bigvee_{a \in \Sigma, \ell' \in \text{Loc}, \psi \in \text{Must}^*(\ell, \ell')} \text{wp}_{\psi}[B(\ell')]
\]

and $\text{predec}^0(B) \equiv B$, $\text{predec}^{j+1}(B) \equiv \text{predec}(\text{predec}^j(B))$ for $j \geq 0$, and then finally $\text{predec}^*(B) \equiv \bigcup_{j \geq 0} \text{predec}^j(B)$. Define $\text{bad} : \text{Loc} \rightarrow \text{Pred}(V^L)$, for any $\ell \in \text{Loc}$, by

\[
\text{bad}(\ell) \equiv \bigvee_{a \in \Sigma, \ell' \in \text{Loc}, \psi \in \text{Must}^*(\ell, \ell')} \exists V^G \cdot \psi \land (\forall(V^L)^\prime, \psi \Rightarrow \bigwedge_{\psi' \in \text{May}^*(\ell, \ell')} \neg \psi')
\]

and thus $\text{bad}(\ell)$ is satisfied by a valuation $s \in [V^L]$ iff there is a must-transition for which no choice of the next data state is permitted by the may-transitions.

In general, for infinite-domain variables, the computation of $\text{predec}^*(\text{bad})$ may not terminate. In [7], it was shown that reachability and related properties in well-structured transition systems with data values, that are monotonic transition systems with a well-quasi ordering on the set of data values, is decidable. This result can be used for specifications with infinite-domain variables to show that under these assumptions, there...
is some $j \geq 0$ such that for all $\ell \in \text{Loc}$, $\llbracket \text{predec}^j(\text{bad})(\ell) \rrbracket = \llbracket \text{predec}^{j+1}(\text{bad})(\ell) \rrbracket$. In the following, for the specification operators conjunction and quotient (which may result in a syntactically inconsistent specification and hence need to be pruned) we assume that such a $j \geq 0$ exists.

The pruning $\rho(A)$ of $A$ is defined if $\varphi^0 \land \neg \text{predec}^j(\text{bad})(\ell)$ is consistent; and in this case, $\rho(A)$ is the specification $(\text{Sig}, \text{Loc}, \ell^0, \varphi^0 \land \neg \text{predec}^j(\text{bad})(\ell^0), E^0, E^\Box)$ where, for $\chi_{\text{good}} = \neg \text{predec}^j(\text{bad}),$

$$E^0 = \{(\ell_1, a, \chi_{\text{good}}(\ell_1) \land \psi \land (\chi_{\text{good}}(\ell_2))', \ell_2) \mid (\ell_1, a, \psi, \ell_2) \in E_0\},$$

$$E^\Box = \{(\ell_1, a, \chi_{\text{good}}(\ell_1) \land \psi \land (\chi_{\text{good}}(\ell_2))', \ell_2) \mid (\ell_1, a, \psi, \ell_2) \in E_\Box\}.$$

Crucially the pruning operator has the expected properties:

**Theorem 3.** Let $A$ be a deterministic, possibly syntactically inconsistent specification. Then $\rho(A)$ is defined if and only if $A$ is consistent. And if $\rho(A)$ is defined, then

1. $\rho(A)$ is a specification (hence syntactically consistent),
2. $\rho(A) \leq A$,
3. $\text{Impl}(A) = \text{Impl}(\rho(A))$, and
4. for any specification $B$, if $B \leq A$, then $B \leq \rho(A)$.

**Logical composition.** Conjunction of two specifications yields the greatest lower bound with respect to modal refinement. Syntactic inconsistencies arise if one specification requires a behavior disallowed by the other.

**Definition 6.** Let $A_1$ and $A_2$ be two specifications with the same signature $\text{Sig}$. The conjunction of $A_1$ and $A_2$ is defined as the possibly syntactically inconsistent specification

$$A_1 \land A_2 = (\text{Sig}, \text{Loc}_1 \times \text{Loc}_2, (\ell_1^0, \ell_2^0), \varphi_1^0 \land \varphi_2^0, E_0, E_\Box)$$

where the transition relations $E_0, E_\Box$ are the smallest relations satisfying the rules, for any $\ell_1 \in \text{Loc}_1, \ell_2 \in \text{Loc}_2, a \in \Sigma$,

1. If $(\ell_1, a, \psi_1, \ell_1') \in E_{0,1}, (\ell_2, a, \psi_2, \ell_2') \in E_{0,2}$, then
   $$((\ell_1, \ell_2), a, \psi_1 \land \psi_2, (\ell_1', \ell_2')) \in E_0,$$
2. If $(\ell_1, a, \psi_1, \ell_1') \in E_{0,1}$, then
   $$((\ell_1, \ell_2), a, \psi_1 \land (\lor_{\psi_2 \in \text{Mag}_2^0(\ell_2, \ell_2')} \psi_2'), (\ell_1', \ell_2')) \in E_0,$$
3. If $(\ell_2, a, \psi_2, \ell_2') \in E_{0,2}$, then
   $$((\ell_1, \ell_2), a, \psi_2 \land (\lor_{\psi_1 \in \text{Mag}_1^0(\ell_1, \ell_1')} \psi_1), (\ell_1', \ell_2')) \in E_0,$$
4. If $(\ell_1, a, \psi_1, \ell_1') \in E_{0,1}$ then
   $$((\ell_1, \ell_2), a, \psi_1 \land (\lor_{\psi_1 \in \text{Mag}_1^0(\ell_1, \ell_1')} \psi_1), (\ell_1', \ell_2')) \in E_0,$$
where $M = \cup_{\ell_1' \in \text{Loc}_1} \text{Mag}_1^0(\ell_1, \ell_1').$
5. If $(\ell_2, a, \psi_2, \ell_2') \in E_{0,2}$ then
   $$((\ell_1, \ell_2), a, \psi_2 \land (\lor_{\psi_1 \in \text{Mag}_1^0(\ell_1, \ell_1')} \psi_1), (\ell_1', \ell_2')) \in E_0,$$
where $M = \cup_{\ell_1' \in \text{Loc}_1} \text{Mag}_1^0(\ell_1, \ell_1').$
The first rule composes may-transitions (with the same action) by conjoining their predicates. Rule (2) and (3) express that any required behavior of $A_1$ ($A_2$ resp.), as long as it is allowed by $A_2$ ($A_1$ resp.), is also a required behavior in $A_1 \land A_2$. Rules (4) and (5) capture the case when a required behavior of $A_1$ is not allowed by $A_2$. Conjunction is commutative and associative.

Refinement is a precongruence with respect to conjunction for deterministic specifications. Moreover, under the assumption of determinism, the conjunction construction yields the greatest lower bound with respect to modal refinement:

**Theorem 4.** Let $A$, $B$, $C$ be specifications with the same signature and let $A$ and $B$ be deterministic. If $A \land B$ is consistent then

1. $\rho(A \land B) \leq A$ and $\rho(A \land B) \leq B$.
2. $C \leq A$ and $C \leq B$ implies $C \leq \rho(A \land B)$.
3. $\text{Impl}(\rho(A \land B)) = \text{Impl}(A) \cap \text{Impl}(B)$.

**Quotient as the dual operator to structural composition.** The quotient operator allows factoring out behaviors from larger specifications. Given two specifications $A$ and $B$ the quotient of $B$ by $A$, in the following denoted $B \Downarrow A$, is the most general specification that can be composed with $A$ and still refines $B$.

In the following, we assume for the signatures $\text{Sig}_A = (\Sigma, V^L_A, V^G_A)$ and $\text{Sig}_B = (\Sigma, V^L_B, V^G_B)$ that $V^L_A \subseteq V^L_B$. The signature of the quotient $B \Downarrow A$ is then $\text{Sig}_{B \Downarrow A} = (\Sigma, V^L_{B \Downarrow A}, V^G_{B \Downarrow A})$ with $V^L_{B \Downarrow A} = V^L_B \setminus V^L_A$ and $V^G_{B \Downarrow A} = V^G_B \cup V^G_A$. Note that, as said before, we restrict ourselves to the case where $V^L_A \cup V^G_A = V^L_B \cup V^G_B$.

It is unknown if in our general model of specifications a finite quotient exists. For specifications involving variables with finite domains only, a semantic quotient operation can be defined, which works on the (finite) semantics of $A$ and $B$. As already noticed in previous works, e.g. [17], non-determinism is problematic for quotienting, and thus specifications are assumed to be deterministic. In our case, even when assuming deterministic specifications, the non-determinism with respect to the next local data state is still there: thus the quotient $B \Downarrow A$, when performing a transition, does not know the next data state of $A$. However, due to our semantics, in which transitions are guarded by uncontrolled states, the quotient can always observe the current data state of $A$. This extension of the usual quotient can be shown that it satisfies the following soundness and maximality property: Given MSD $S$ and $T$ such that $S$ is deterministic and $T \Downarrow_{\text{sem}} S$ is consistent, and assume a semantic pruning operator $\rho_{\text{sem}}$ which is the straightforward translation of pruning $\rho$ to the semantic level. Then $X \leq_{\text{sem}} \rho_{\text{sem}}(T \Downarrow_{\text{sem}} S)$ if and only if $S \models_{\text{sem}} X \leq_{\text{sem}} T$ for any MSD $X$.

Now our goal is to compute the quotient at the symbolic level of specifications. We do this for a restricted subclass of specifications in which each occurring transition predicate $\psi$ is separable, meaning that $\psi$ is equivalent to $\circ \psi \land \psi'$. Although this might seem as a serious restriction, we can often transform the transition systems with transition predicates of the form $(x)' = x + 1$ to a transition system with transition predicates which are separable and keep the same set of implementations. For instance, if we know that there are only finitely many possible values $v_1, \ldots, v_n$ for $x$ in the current state, we can “unfold” the specification and replace the transition predicates $(x)' = x + 1$ by $(x)' = v_i$, for $1 \leq i \leq n$. 

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The symbolic quotient introduces two new locations, the universal state (univ) and an error state (∐). In the universal state the quotient can show arbitrary behavior and is needed to obtain maximality, and the error state is a syntactically inconsistent state used to encode the quotient is giving requirements. The state space of the quotient is given by \( \text{Loc}_B \times \text{Loc}_A \times \text{Pred}(V^L_A) \), so every state stores not only the current location of \( B \) and \( A \) (like in [17]) but includes a predicate about the current possible data states of \( A \). For notational convenience, for \( \varphi \in \text{Pred}(V_1 \cup V_2) \) and \( \varphi_1 \in \text{Pred}(V_1) \), we write \( \varphi \parallel \varphi_1 \) for \((\forall \varphi_1. \varphi_1 \Rightarrow \varphi) \in \text{Pred}(V_2)\).

**Definition 7.** Let \( A \) and \( B \) be two specifications such that \( V^L_A \subseteq V^L_B \). The quotient of \( B \) by \( A \) is defined as the possibly syntactically inconsistent specification \( B \parallel A = (\text{Sig}_B \setminus A) \cup (\text{Loc}_B \times \text{Loc}_A \times \text{Pred}(V^L_A)) \cup \{ \text{univ} \}, (\ell^0_B, \ell^A_A, \varphi^A_A), \varphi^B_B, \varphi^A_A \parallel \varphi^B_B, E_0, E_\square \) where the transition relations are given by, for all \( a \in \Sigma \) and all \( \ell \) in \( \text{Pred}(V^L_A) \),

1. if \( (\ell_B, a, \psi_B, \ell^A_A) \in E_{0,B} \) and \( (\ell_A, a, \psi_A, \ell^B_B) \in E_{0,B} \), then
   \( ((\ell_B, \ell_A, \xi_A), a, \xi_A \land a \psi_B \land \neg a \psi_A \land (\psi^B_B \cup \psi^A_A)) \in E_{0} \),
2. if \( (\ell_B, a, \psi_B, \ell^A_A) \in E_{\square,B} \) and \( (\ell_A, a, \psi_A, \ell^B_B) \in E_{\square,B} \), then
   \( ((\ell_B, \ell_A, \xi_A), a, \xi_A \land a \psi_B \land \neg a \psi_A \land (\psi^B_B \cup \psi^A_A)) \in E_{\square} \),
3. if \( (\ell_B, a, \psi_B, \ell^A_A) \in E_{\square,B} \) and \( (\ell_A, a, \psi_A, \ell^B_B) \in E_{\square,A} \), then
   \( ((\ell_B, \ell_A, \xi_A), a, \xi_A \land a \psi_B \land \neg a \psi_A \land (\psi^B_B \cup \psi^A_A)), \bot \) \( \in E_{\square} \),
4. if \( (\ell_B, a, \psi_B, \ell^A_A) \in E_{\square,B} \),
   \( ((\ell_B, \ell_A, \xi_A), a, \xi_A \land a \psi_B \land \neg a \psi_A \land (\psi^B_B \cup \psi^A_A)) \in E_{\square} \)
   where \( M = \bigcup_{a \in \Sigma} \text{Loc}_A \text{Must}_A(\ell_A, \ell^A_A) \),
5. if \( (\ell_B, \ell_A, \xi_A), a, \neg \psi_A, \text{univ} \) \( \in E_0 \),
6. if \( (\ell_B, \ell_A, \xi_A), a, \xi_A \land \neg (\forall \psi_A \in M \cdot \psi_A), \text{univ} \) \( \in E_0 \)
   where \( M = \bigcup_{a \in \Sigma} \text{Loc}_A \text{May}_A(\ell_A, \ell^A_A) \),
7. \((\text{univ}, a, \text{true}, \bot) \in E_0 \),
8. \((\bot, a, \text{true}, \bot) \in E_{\square} \).

Rules (1) and (2) capture the cases when both \( A \) and \( B \) can perform a may- and must-transition, respectively. Rules (3) and (4) capture any inconsistencies which can arise if for a must-transition in \( B \) there is no way to obtain a must-transition by composition of the quotient with \( A \). In order to obtain maximality, we add a universal state univ in which the behavior of the quotient is not restricted (rules (5)–(7)). Finally, the rule (8) makes the error state syntactically inconsistent.

Since we only have finitely many transition predicates \( \psi_A \) in \( A \), and they are all separable, the set of locations \( \text{Loc}_B \times \text{Loc}_A \times \{(\psi^A_A | \psi_A \text{ occurring in } A) \cup \{\psi^B_B\}) \cup \{\text{univ}, \bot\} \) of \( B \parallel A \) is also finite. Thus we can construct the symbolic quotient in a finite number of steps, starting in the initial state \((\ell^0_B, \ell^A_A, \varphi^A_A)\), and iteratively constructing the transitions. Soundness and maximality of the quotient follows from the following theorem.

**Theorem 5.** Let \( A \) and \( B \) be specifications such that \( V^L_A \subseteq V^L_B \), all transition predicates of \( A \) and \( B \) are separable, \( A \) is deterministic and \( B \parallel A \) is consistent. Then for any specification \( C \) such that \( \text{Sig}_C = \text{Sig}_B \setminus A \), \( C \leq \rho(B \parallel A) \) if and only if \( A \parallel C \leq B \).
4 Predicate Abstraction for Verification of Refinement

We now switch our focus to the problem of deciding whether a specification $A$ refines
another specification $B$ (which reduces to checking $(A)_{sem} \leq_{sem} (B)_{sem}$). As soon
as domains of variables are infinite, $(A)_{sem}$ and $(B)_{sem}$ may be MSD with infinitely
many states and transitions. In this case, this problem is known to be undecidable in
general. Thus we propose to resort to predicate abstraction techniques [18]. Given two
specifications $A$ and $B$ we derive over- and under-approximations $A^o$ and $B^u$ which
are guaranteed to be finite MSD. Then, we show that $A^o \leq_{sem} B^u$ implies $A \leq B$.

Example 5. Fig. 5 shows a print server specification $Q$ which we will show is a re-
finement of the abstract specification $P$ in Fig. 1. The behavior of the print server is
now fixed for any number of print jobs. Moreover, the send transition has been refined
such that depending on the priority mode (provided by the environment of the print
server) a job with high priority (in case $priorityMode$ is true) or a job with low priority
(otherwise) is chosen next.

Given a specification $A = (Sig, Loc, \ell^0, \rho^0, \rightarrow_0, \rightarrow_G)$ with $Sig = (\Sigma, V^L,
V^G)$, we partition the local state space and the uncontrolled state space using finitely
many predicates $\phi_1, \phi_2, \ldots, \phi_N \in Pred(V^L)$ and $\chi_1, \chi_2, \ldots, \chi_M \in Pred(V^G)$. We
fix these predicates in the following to simplify the presentation. The signature of the
abstraction is then given by $Sig_{abstr} = (\Sigma, V^L_{abstr}, V^G_{abstr})$, where $V^L_{abstr} = \{x_1, x_2,
\ldots, x_N\}$ and $V^G_{abstr} = \{y_1, y_2, \ldots, y_M\}$. All variables $x_i, y_j$ have Boolean domain. A
variable $x_i$ (resp. $y_j$) encodes whether the predicate $\phi_i$ ($\chi_j$) holds or not.

Any abstract state $\nu \in [V^L_{abstr}]$ is a conjunction of predicates $\bigwedge_{i=1}^N \phi_i^{\nu(x_i)}$, where $\phi_i^{\nu(x_i)} = \phi_i$ if $\nu(x_i) = 1$, else $\phi_i^{\nu(x_i)} = \neg \phi_i$. Further, a set of abstract states $N \subseteq [V^L_{abstr}]$
corresponds to $\bigvee_{\nu \in N} \nu$. Similarly for any $\omega \in [V^G_{abstr}]$ and for $M \subseteq [V^G_{abstr}]$.

The transition relation of the over-approximation expands the allowed behaviors and
limits the required behaviors. Dually, the under-approximation will further restric-
the allowed behavior and add more required transitions. In other words, over-
approximation is an existential abstraction on may-transitions and universal abstraction
on must-transitions; dually for the under-approximation.

Formally, the over-approximation $A^o$ of $A$ is defined by the finite TSD $\langle \tilde{Sig}_{abstr}, \tilde{Loc}, \tilde{\ell}^0, \tilde{S}^0_{abstr}, \rightarrow_0, \rightarrow_G, \rightarrow_\omega, \rightarrow_\chi \rangle$, where the initial abstract state contains all partitions overlapping with concrete initial states $S^0_{abstr} = \{\nu \in [V^L_{abstr}] \mid \exists \ell^0, \nu \land \rho^0\}$.
and the abstract transition relations are derived as follows. For all $\ell, \ell' \in \text{Loc}$, $a \in \text{Act}$, $\nu, \dot{\nu} \in \llbracket \text{VL}_{\text{abstr}} \rrbracket$, we have:

1. If $\exists V. (V^L)' \land \omega \land (\forall \dot{\psi} \in \text{May}_{\ell}(\ell, \ell')) \land (\dot{\nu})'$, then $(\ell, \nu) \xrightarrow{\omega_a}_{\text{□}, \text{abstr}} (\ell', \{\dot{\nu}\})$.
   
   so there is a may-transition between partitions in the abstraction if there was a may-transition between any states in these partitions in the concrete system.

2. Whenever, for some $N \subseteq \llbracket V^L_{\text{abstr}} \rrbracket$, the predicate
   
   $$\forall \nu \land \omega \Rightarrow \forall \psi \in \text{Must}_{\ell}(\ell, \ell') \circ \psi \land (\forall (V^L)' \land \psi \Rightarrow (N)' \land (N)') \quad (2)$$

   is true and $N$ is minimal with respect to this property, then $(\ell, \nu) \xrightarrow{\omega_a}_{\text{□}, \text{abstr}} (\ell', N)$.

For the under-approximation $B^u$ of $B$, we assume that every transition predicate $\psi$ on a must-transition must be separable (see page 12). Moreover, in order to soundly capture must-transitions, we must be able to exactly describe the target set of (concrete) local states by a union of abstract states; so for any $(\ell, a, \psi, \ell') \in E_{\square, B}$, there exists a set $N \subseteq \llbracket V^L_{\text{abstr}} \rrbracket$ such that $\forall (V^L)' \land \psi \Rightarrow (N)'$. The under-approximation $B^u$ is the finite TSD $(\text{Sig}_{\text{abstr}}, \text{Loc}, \ell_0, S_{\text{abstr}}^0, \rightarrow_{\text{□}, \text{abstr}}, \rightarrow_{\square, \text{abstr}})$, where $S_{\text{abstr}}^0 = \{ \nu \in \llbracket V^L_{\text{abstr}} \rrbracket \mid \forall V. \nu \Rightarrow \psi^{N} \}$, and for all $\ell, \ell' \in \text{Loc}$, $a \in \text{Act}$, $\nu, \dot{\nu} \in \llbracket V^L_{\text{abstr}} \rrbracket$, we have:

i. If $\forall V. (V^L)' \land \omega \land (\dot{\nu})' \Rightarrow \forall \psi \in \text{May}_{\ell}(\ell, \ell') \psi$ then $(\ell, \nu) \xrightarrow{\omega_a}_{\square, \text{abstr}} (\ell', \{\dot{\nu}\})$.

ii. For every $(\ell, a, \psi, \ell') \in E_{\square, B}$, if $\exists V. \nu \land \omega \land \psi \Rightarrow (N)'$, then $(\ell, \nu) \xrightarrow{\omega_a}_{\text{□}, \text{abstr}} (\ell', N)$ where $N \subseteq \llbracket V^L_{\text{abstr}} \rrbracket$ such that $\forall (V^L)' \land \psi \Rightarrow (N)'$.

Correctness of the abstraction follows from the following theorem.

**Theorem 6.** $A^o \leq_{\text{sem}} B^u$ implies $A \leq B$.

**Example 6.** Fig. 6 and Fig. 7 are over- and under-approximations of $Q$ and $P$, respectively. The MSD represent abstractions w.r.t. the predicates $\phi_{0, 0} \equiv h = l = 0$, $\phi_{0, 1} \equiv l = 0 \land h = 1$, $\phi_{1, 0} \equiv l = 1 \land h = 0$, and $\phi_{2, 1} \equiv h + l > 1$ for the controlled variables $l$ and $h$, and $\omega_1 \equiv \text{priorityMode}$, $\omega_2 \equiv \neg \text{priorityMode}$ for the uncontrolled variable $\text{priorityMode}$. Note that all transition predicates in $P$ are separable, and all possible (concrete) poststates can be precisely captured by the predicates $\phi_{0, 0}, \phi_{0, 1}, \phi_{1, 0}, \phi_{2, 1}$. For better readability we have omitted most of the guards $\omega_1, \omega_2$, i.e., every transition without guard stands for two transitions with the same action, source and target state(s), and with $\omega_1$ and $\omega_2$ as guard, respectively. Moreover, the state $(\ell_3, \phi_{0, 0} \land \phi_{0, 1} \land \phi_{1, 0} \land \phi_{2, 1})$ is a simplified notation which represents all the states $(\ell_3, \phi)$ with $\phi \in \{\phi_{0, 0}, \phi_{0, 1}, \phi_{1, 0}, \phi_{2, 1}\}$ and all may-transitions leading to it lead to each of the states, and the may-loop stands for all the transitions between each of the states. Obviously, $Q^o \leq_{\text{sem}} P^u$, and from Thm. 6 it follows that $Q \leq P$.

Even though this abstraction technique requires separability of predicates, it is applicable to a larger set of specifications. Sometimes, as already described in the previous section, transitions with non-separable predicates can be replaced by finite sets of transitions to achieve separability, without changing the semantics of the specification. Automatic procedures for generation of predicates are subject of future work. Finally, our abstraction also supports compositional reasoning about parallel composition in the following sense:
Theorem 7. Let A and B be two composable specifications, and $V_A^G = (V_A^L \cup V_B^L) \setminus (V_A^L \cup V_B^L)$. Let $E_A \subseteq \text{Pred}(V_A^L)$, $E_B \subseteq \text{Pred}(V_B^L)$, and $F \subseteq \text{Pred}(V_A^G \cup V_B^G)$ be sets of predicates partitioning the respective data states.

A is approximated w.r.t. $E_A$ for $V_A^L$, and $E_B \cup F$ for $V_A^G = V_A^L \cup V_B^L$ and similarly, B is approximated w.r.t. $E_B$ and $E_A \cup F$. Finally, $A \parallel B$ is approximated w.r.t. $E_A \cup E_B$ for $V_A^L \parallel V_B^L$, and $F$ for $V_A^G \parallel V_B^G$. We assume that each predicate, in any abstraction of A, B, or $A \parallel B$, are encoded with the same variable.

Then $(A \parallel B)^o \leq_{\text{sem}} A^o \parallel_{\text{sem}} B^o$, and $A^u \parallel_{\text{sem}} B^u \leq_{\text{sem}} (A \parallel B)^u$.

This result allows reusing abstractions of individual components in a continued development and verification process. For instance, if we want to verify $A \parallel B \leq C$ then we can compute (or reuse) the less complex abstractions $A^o$ and $B^o$. Thm. 7 implies then that from $A^o \parallel_{\text{sem}} B^o \leq_{\text{sem}} C^u$ we can infer $A \parallel B \leq C$.

5 Related work

The main difference to related approaches based on modal process algebra taking data states into account, e.g. [19] is that they cannot naturally express logical and structural composition in the same formalism. A comparison between modal specifications and
other theories such as interface automata [20] and process algebra [2] can be found in [3]. In [8], the authors introduced sociable interfaces, that is a model of I/O automata [21] equipped with a data and a game-based semantics. While their communication primitives are richer, sociable interfaces do not encompass any notion of logical composition and quotient, and their refinement is based on an alternating simulation.

Transition systems enriched with predicates are used, for instance, in the approach of [22, 23] where they use symbolic transition systems (STS), but STS do not support modalities and loose data specifications as they focus more on model checking than on the (top down) development of concurrent systems by refinement.

In [15] modal I/O automata has been extended by pre- and postconditions viewed as contracts, however, only semantics in terms of sets of implementations have been defined (implementations with only input actions correspond to our TSD). Modal refinement as defined in [15] is coarser than in this paper, and moreover, neither conjunction nor a quotient operation are defined.

6 Conclusion

We have proposed a specification theory for reasoning about components with rich data state. Our formalism, based on modal transition systems, supports: refinement checking, consistency checking with pruning of inconsistent states, structural and logical composition, and a quotient operator. We have defined symbolic representations of the operators and have shown that they are equivalent to the semantic definitions—this allows for automatic analysis of specifications. We have also presented a predicate abstraction technique for modal specifications with data. We believe that this work is a significant step towards practical use of specification theories based on modal transition systems. The ability to reason about data domains permits the modeling of industrial case studies.

In the future, we intend to develop larger case studies. Furthermore, we would like to extend the formalism with more complex communication patterns and to investigate in which cases we can still obtain all the operators on specifications, in particular the quotient operator. We are also planning to implement the theory in the MIO Workbench [24, 25, 26], a verification tool for modal input/output interfaces.

References


[26] MIO Workbench: http://www.miowb.net/
Evaluating the performance of model transformation styles in Maude*

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Abstract. Rule-based programming has been shown to be very successful in many application areas. Two prominent examples are the specification of model transformations in model driven development approaches and the definition of structured operational semantics of formal languages. General rewriting frameworks such as Maude are flexible enough to allow the programmer to adopt and mix various rule styles. The choice between styles can be biased by the programmer’s background. For instance, experts in visual formalisms might prefer graph-rewriting styles, while experts in semantics might prefer structurally inductive rules. This paper evaluates the performance of different rule styles on a significant benchmark taken from the literature on model transformation. Depending on the actual transformation being carried out, our results show that different rule styles can offer drastically different performances. We point out the situations from which each rule style benefits to offer a valuable set of hints for choosing one style over the other.

1 Introduction

Many engineering activities are devoted to manipulate software artifacts to enhance or customize them, or to define their possible ordinary evolutions and exceptional reconfigurations. The concept of model as unifying software artifact representation has been promoted as a means to facilitate the specification of such activities in a generic way. Many dynamic aspects can be conceived as model transformations: e.g. architectural reconfigurations, component adaptations, software refactorings, and language translations. Rule-based specifications have been widely adopted as a declarative approach to enact model-driven transformations, thanks to the intuitive meaning and solid foundations offered by rule-based machineries like term [1] and graph rewriting [2].

Recently we have investigated the possibility to exploit the structure of models to enhance software description and to facilitate model transformations [3, 4]. Indeed, many domains exhibit an inherently hierarchical structure that can be exploited conveniently to guarantee scalability. We mention, among others, nested components in software architectures and reflective object-oriented systems, nested sessions and transactions in business processes, nested membranes

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in computational biology, composition associations in UML-like modeling frameworks, semi-structured data in XML-like formats, and so on. Very often such layering is represented in a plain manner by overlapping the intra- and the inter-layered structure. For instance, models are usually formalised as flat object configurations (e.g. graphs) and their manipulation is studied with tools and techniques based on rewriting theories that do not fully exploit the hierarchical structure. On the other hand, an explicit treatment of the hierarchical structure for specifying and transforming model-based software artifacts is possible. As a matter of fact, some layering structures (like composition relations in UML-like languages) can be conveniently represented by an explicit hierarchical structure enabling then hierarchical manipulations of the resulting models (see e.g. [3, 4]).

We have investigated such issues in previous work [3] proposing an approach analogous to the Russian dolls of [5, 6], where objects can be nested within other objects. In this view, structured models are represented by terms that can be manipulated by means of term-rewrite techniques like conditional term rewriting [1]. In [3] we compared the flat representation against the nested one, showing that they are essentially equivalent in the sense that one can bijectively pass from one to the other. Each representation naturally calls for different rule styles and the comparison in [3] mainly addressed methodological aspects, leaving one pragmatical issue open: how to decide in advance which approach is more efficient for actually executing a model transformation?

We offer an answer to this question in this paper. We have selected two prominent approaches to model transformation. The first one is archetypal of the graph-transformation based model-driven community and follows the style of [7]. The second one is quite common in process calculi and goes along the tradition of Plotkin’s structural operational semantics, as outlined in [3]. Both approaches can be adopted in flexible rule-based languages like Maude [8] (the rewriting logic based language and framework we have chosen). In order to obtain significant results we have implemented three test cases widely used in the literature: the reconfiguration of components that migrate from one location to another one, the transformation of class diagrams into relational schemas, and the refactoring of class diagrams by pulling up attributes. As a byproduct we offer a novel implementation of these three classical transformations based on conditional rules. Indeed, such style of programming model transformations has not been proposed by other authors, as far as we know.

Our experimental results stress the importance of choosing the right transformation style carefully to obtain the best possible performance. We point out some features of the examples that impact on the performance of each rule format, thus providing the programmer with a set of valuable guidelines for programming model transformations in expressive rule-based frameworks like Maude.

Synopsis. § 2 presents a graph-based algebraic representation of models as nested object collections and describes rewrite rule styles for implementing model transformations in Maude. § 3 presents some enhancements that can be applied to the transformation styles. § 4 describes our benchmark. § 5 presents the experimental results. § 6 concludes the paper.
2 Preliminaries

In this section we illustrate the two key model transformation paradigms and the Maude notation we shall exploit in the rest of the paper over a basic example of transformation, namely from trees to list. A classical approach would provide ad-hoc data structures for trees and lists and an ad-hoc algorithm for implementing the transformation. Model driven approaches, instead, consider a common representation formalism for both data structures and a generic transformation procedure that acts on such formalism. In our setting, the representation formalism for models are collections of attributed objects and the transformation procedure is based on rewrite rules.

The Maude language already provides some machinery for this purpose, called object-based configurations [8], which we tend to follow with slight modifications aimed to ease the presentation. More precisely we represent models as nested object collections [3] (following an idea originally proposed in [5] and initially sketched in [6]), which can be understood as a particular class of attributed, hierarchical graphs. We then implement transformations as sets of rewrite rules.

Rewriting Logic and Maude. Maude modules describe theories of rewriting logic [1], which are tuples \( \langle \Sigma, E, R \rangle \) where \( \Sigma \) is a signature, specifying the basic syntax (function symbols) and type machinery (sorts, kinds and subsorting) for terms, e.g. model descriptions; \( E \) is a set of (possibly conditional) equations, which induce equivalence classes of terms, and (possibly conditional) membership predicates, which refine the typing information; \( R \) is a set of (possibly conditional) rules, e.g. model transformations.

The signature \( \Sigma \) and the equations \( E \) of a rewrite theory form a membership equational theory \( \langle \Sigma, E \rangle \), whose initial algebra is denoted by \( T_{\Sigma/E} \). Indeed, \( T_{\Sigma/E} \) is the state space of a rewrite theory, i.e. states (e.g. models) are equivalence classes of \( \Sigma \)-terms modulo the least congruence induced by the axioms in \( E \) (denoted by \([t]_E\) or \( t \) for short). Sort declarations takes the form sort \( S \) and subsorting (i.e. subtyping) is written \( \text{subsort } S < T \). For instance, the sort of objects (\( \text{Obj} \)) is a subsort of configurations sort \( \text{Configuration} \) as declared by \( \text{subsort } \text{Obj} < \text{Configuration} \).

Operators are declared in Maude notation as op \( f : \text{TL} \rightarrow T \text{ [As]} \) where \( f \) is the operator symbol (possibly with mixfix notation where underscores stand for argument placeholders), \( \text{TL} \) is a (possibly empty, blank separated) list of domain sorts, \( T \) is the sort of the co-domain, and \( \text{As} \) is a set of equational attributes (e.g. associativity, commutativity). For example, object configurations (sort \( \text{Configuration} \)) are constructed with operators for the empty configuration (none: -> \( \text{Configuration} \)), single objects (via subsorting) or the union of configurations, denoted with juxtaposition (\( . : \text{Configuration} \rightarrow \text{Configuration} \text{ [assoc comm id:none]} \), declared to be associative, commutative and to have none as its identity operator (i.e. they are multisets).

Each object represents an entity and its properties. Technically, an object is defined by its identifier (of sort \( \text{Oid} \)), its class (of sort \( \text{Cid} \)) and its attributes (of
sort AttSet). Objects are built with an operation < : | > with functional type Oid Cid AttSet -> Obj. Following Maude conventions, we shall use quoted identifiers like 'a as object identifiers, while class identifiers will be defined by ad-hoc constructors. In our running example we use the constants Node and Item of sort Cid to denote the classes of tree nodes and list items, respectively.

The attributes of an object define its properties and relations to other objects. They are basically of two kinds: datatype attributes and association ends. Datatype attributes take the form n: v, where n is the attribute name and v is the attribute value. For instance, in our running example we shall consider a natural attribute value (sort Nat), representing the value of a node or item. A node with identifier 'a and value 5 is denoted by < 'a : Node | value: 5 >.

Relations between objects can be represented in different ways. One typical approach is to use a pair of references (called association ends in UML terminology) for each relation. So if an object o1 is in relation R with object o2 then o1 is equipped with a reference to o2 and vice versa. In our case this is achieved with attributes of the form R: O2 and opp(R): O1 where R indicates the relation name and O1, O2 are sets of object identifiers (sort OidSet). Association ends of the same relation within one object are grouped together (hence the use of identifier sets as domain of association attributes). In our example we have two relations left and right between a node and its left and right children, and one relation next between an item of the list and the next one. Clearly, the opposite relations of left, right and next are the parent and previous relations. As an example of a pair of references consider a node < 'a : Node | value: 5 , left: 'b > and its son < 'b : Node | value: 3 , opp(left): 'a >. Of course an object can be equipped with any number of attributes. Actually, the attributes of an object form a set built out of singleton attributes, the empty set (none) and union set (denoted with | |).

The following simple configuration represents a tree with three nodes.

< 'a : Node | value: 5 , left: 'b , right: 'c >
< 'b : Node | value: 3 , opp(left): 'a >
< 'c : Node | value: 7 , opp(right): 'a >

Operation << > >> : Configuration -> Model wraps a configuration into a model.

Functions (and equations that cannot be declared as equational attributes) are defined by a set of confluent and terminating conditional equations of the form ceq t = t' if c, where t, t' are Σ-terms, and c is an application condition. When the application condition is vacuous, the simpler syntax eq t = t' can be used. For example, an operator op size : Configuration -> Nat for measuring the number of objects in a configuration is inductively defined by equations eq size(none) = 0 and eq size(O C) = 1 + size(C) (with O, C being variables of sort Obj, Configuration, respectively). Roughly, an equational rule can be applied to a term t' if we find a match m (i.e. a variable substitution) for t at some place in t'' such that m(c) holds (i.e. c after the application of the substitution m evaluates to true). The effect is that of substituting the matched part with m(t'). For example, calculating the size of the above tree is done by...
reducing $size(< \text{a} : \text{Node} | \text{value: 5} , \text{left: b} , \text{right: c} > < \text{b} : \text{Node} | \text{value: 3} , \text{opp(left): a} > < \text{c} : \text{Node} | \text{value: 7} , \text{opp(right): a} >)$ to $1 + size(< \text{b} : \text{Node} | \text{value: 3} , \text{opp(left): a} > < \text{c} : \text{Node} | \text{value: 7} , \text{opp(right): a} >)$, then to $2 + size(< \text{c} : \text{Node} | \text{value: 7} , \text{opp(right): a} >)$ and finally to $3$.

Structured models. A nested object collection allows objects to have container attributes, i.e. configuration domain attributes. While in a plain object collection a containment relation $r$ between two objects $o_1$ and $o_2$ is represented by exploiting a pair of association end attributes $r$ and $\text{opp}(r)$, now $o_2$ is embedded into $o_1$ by means of the container attribute $r$. For instance, the above tree becomes

$$< \text{a} : \text{Node} | \text{value: 5} ,$$
$$\quad \text{left: } < \text{b} : \text{Node} | \text{value: 3} > ,$$
$$\quad \text{right: } < \text{c} : \text{Node} | \text{value: 7} > >$$

The hierarchical structure of models forms a tree. The two approaches that we have described differ essentially in the way we represent such a tree. Indeed, flat and nested representations are in bijective correspondence, i.e. for each flat object collection we can obtain a unique nested collection and vice versa as shown in [3], so that we can pass from one to the other as we find more convenient for specific applications or analyses.

Transformations as sets of rewrite rules. Transformations can be defined by means of rewrite rules, which take the form $crl \ t \Rightarrow t'$ if $c$, where $t$, $t'$ are $\Sigma$-terms, and $c$ is an application condition (a predicate on the terms involved in the rewrite, further rewrites whose result can be reused, membership predicates, etc.). When the application condition is vacuous, the simpler syntax $rl \ t \Rightarrow t'$ can be used. Matching and rule application are similar to the case of equations with the main difference being that rules are not required to be confluent and terminating (as they represent possibly non-deterministic concurrent actions rather than functions). Equational simplification has precedence over rule application in order to simulate rule application modulo equational equivalence.

SPO transformations. The need for visual modelling languages and the graph-based nature of models have contributed to the success of graph transformation approaches to model transformations. In such approaches, transformations are programmed in a declarative way by means of a set of graph rewrite rules. The transformation style that we consider here is based on the algebraic graph transformation approach [2]. The main idea is that each rule has a left-hand side and a right-hand side pattern. Each pattern is composed by a set of objects (nodes) possibly interrelated by means of association ends (edges). A rule can be applied to a model whenever the left-hand side can be matched with part of the model, i.e. each object in the left-hand side is (injectively) identified with an object and idem for the association ends. The application of a rule removes the matched part of the model that does not have a counterpart in the right-hand side and, vice versa, adds to the model a fresh copy of the right-hand side part.
that is not present in the left-hand side. Items in common between the left-hand side and the right-hand side are preserved during the application of the rule. Very often, rules are equipped with additional application conditions, including those typical of graph transformation systems (e.g. to avoid dangling edges) and its extensions like Negative Application Conditions (NACs).

In our setting, this means that rules have in general the following format:

\[
\text{crl} \leftarrow \text{lhs conf1} \rightarrow \text{rhs} \text{ conf1} \iff \text{applicable(lhs conf1)}.
\]

where \( \text{lhs} \) and \( \text{rhs} \) stand for the rule’s left- and right-hand side configurations, \( \text{conf1} \) as the context in which the rule will be applied, and \( \text{applicable} \) is the boolean function implementing the application condition. Simpler forms are possible, e.g. in absence of application conditions the context is not necessary and rules take the form: \( \text{rl } \text{lhs} \rightarrow \text{rhs} \).

In our running example the transformation rules basically take a node \( x \) and its children \( y \) and \( z \) and puts them in some sequence, with \( x \) before \( y \) and \( z \). This rule might introduce branches in the sequence that are solved by appropriate rules. A couple of rules are needed to handle some special cases, like \( x \) being the root or a node that has already been put in the list (in the middle, tail or head). Let us show one of the basic rules (the rest of the rules are very similar):

\[
\text{rl [nodeToItem]} \leftarrow \text{<< < x : Node | value: vx , left: y , right: z , next: u , Ax > \n y : Node | value: vy , op(left): x, Ay > \n z : Node | value: vz , op(right): x, Az > \n u : Node | value: vu , op(next): x, Au > \n conf1 >> \rightarrow \text{<< < x : Item | value: vx , next: y , Ax> \n y : Node | value: vy , op(next): x, next: z , Ay > \n z : Node | value: vz , op(next): y, next: u , Ay > \n u : Node | value: vu , op(next): z, Au > \n conf1 >>}.
\]

**SOS transformations.** We now describe transformation rules in the style of Structural Operational Semantics [9] (SOS). The basic idea is to define a model transformation by structural induction, which in our setting basically amounts to exploiting set union and (possibly) nesting.

We recall that SOS rules make use of labels to coordinate rule applications. We first present the implementation style of SOS semantics in rewriting logic as described in [10] and then present our own encoding of SOS which provides a more efficient implementation, though circumscribed to some special cases.

The approach of [10] requires to enrich the signatures with sorts for rule labels \( \text{Lab} \), label-prefixed configurations \( \text{LabConfiguration} \), and a constructor \( \{ \} \) : \( \text{Lab Configuration} \Rightarrow \text{LabConfiguration} \) for label-prefixed configurations. In addition, rule application is allowed at the top-level of terms only (via Maude’s \textit{frozen} attribute [11]) so that sub-terms are rewritten only when required in the premise of a rule (as required by the semantics of SOS rules). With this notation
a term \{lab1\}conf1 represents that a configuration conf1 has been obtained after application of a lab1-labelled rule.

One typical rule format in our case allows us to conclude a transformation lab1 for a configuration made of two parts conf1 and conf2 provided that each part can respectively perform some transformation lab2, lab3:

\[
\text{crl conf1 conf2 } \to \text{ \{lab3\} conf3 conf4} \\\text{if conf1 } \to \text{ \{lab1\} conf3} \\\text{\lor conf2 } \to \text{ \{lab2\} conf4} .
\]

Typically, the combination of labels will follow some classical form. For instance, with Milner-like synchronisation, lab1, lab2 can be complementary actions, in which case lab3 would be a silent action label. Instead, Hoare-like synchronisation would require lab1, lab2 and lab3 to be equal.

Consider now a hierarchical representation of models based on nested object collections. In this situation we need rules for dealing with nesting. Typically, the needed rule format is the one that defines the transformation lab1 of an object oid1 conditional to some transformation lab2 of one of its contents c:

\[
\text{crl < oid1 : cid1 | c: conf1 , attSet1 > } \to \text{ \{lab1\} < oid1 : cid1 | c: conf2 , f(attSet1) >} \\\text{if conf1 } \to \text{ \{lab2\} conf2} .
\]

Such rules might affect the attributes of the container object (denoted with function f) but will typically not change the object’s identifier or class. More elaborated versions of the above rule are also possible, for instance involving more than one object or not requiring any rewrite of contained objects.

In our running example we have the following rule that transforms a tree provided that its subtrees can be transformed into lists

\[
\text{crl [root] : < x : Node | value: vx , left: leftTree , right: rightTree > } \to \text{ \{toList\}}
\]

\[
\text{list1} \\\text{if leftTree } \to \text{ \{toList\} list1} \\\text{< tail : Item | value: vt , opp(next): y >} \\\text{< x : Item | value: vx , opp(next): tail , next: head >} \\\text{< head : Item | value: vh , opp(next): x , next: z >} \\\text{list2} \\\text{\lor rightTree } \to \text{ \{toList\} < tail : Item | value: vt , opp(next): y >} \\\text{< head : Item | value: vh , next: z > list2} .
\]

Note that head and tail of the transformed sublists are identified by the lack of next and opp(next) attributes. Rules are also needed to handle leafs:

\[
\text{rl [root] : < x : Node | value: vx > } \to \text{ \{toList\} < x : Node | value: vx >} .
\]

Finally, rules are needed to close the transformations at the level of models. Such rules have the following format:

\[
\text{crl << conf1 >> } \to \text{ \{toList\} conf2} \\\text{if conf1 } \to \text{ \{toList\} conf2} .
\]

In our example the rule would be

\[
\text{crl << conf1 >> } \to \text{ \{toList\} conf2} \\\text{if conf1 } \to \text{ \{toList\} conf2} .
\]

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3 Enhanced SOS implementation

While performing our preliminary experiments we discovered a more efficient way to encode SOS rules in rewriting logic that we call SOS*.

The most significant improvement applies to those cases in which the labels of the sub-configurations are known in advance. As a matter of fact this was the case of all test cases we consider in the next section. The idea is to put the labels on the left-hand side of rules as a sort of context requiring the firing of transformations with such label. In other words, we pass from post- to pre-rule applicability checks.

As a more general example the above rule scheme becomes now:

\[
\begin{align*}
crl \{lab3\} & \text{ conf1 conf2 } \Rightarrow \text{ conf3 conf4} \\
\text{if } & \{lab1\} \text{ conf1 } \Rightarrow \text{ conf3} \\
\land & \{lab2\} \text{ conf2 } \Rightarrow \text{ conf4}.
\end{align*}
\]

The main difference is that now lab2 and lab3 are known in advance and not obtained as a result of the conditional rewrites. A notable example where this alternative encoding cannot be immediately applied are the semantics of process calculi where synchronisation rules do not know in advance which signals are ready to perform their subprocesses.

Another slight improvement is the object-by-object decomposition of object collections instead of the one based on a pair of subsets presented above. For example the above rule scheme becomes:

\[
\begin{align*}
crl \{lab1\} & \text{ obj1 conf2 } \Rightarrow \text{ obj3 conf4} \\
\text{if } & \{lab2\} \text{ obj1 } \Rightarrow \text{ obj3} \\
\land & \{lab3\} \text{ conf2 } \Rightarrow \text{ conf4}.
\end{align*}
\]

A more significant improvement is that in some cases we allow to contextualise rules at any place of a term. We recall that in a SOS derivation this is typically achieved by rules that lift up silent (e.g. \(\tau\)-labelled) actions. Technically this is essentially achieved by declaring as frozen the labelling operator \(\{\_\}\) only. This allows to apply rules to transform a sub-configuration at any level of the nesting hierarchy. That is, SOS rules like the ones for lifting silent actions across the nesting hierarchy like

\[
\begin{align*}
crl < oid1 : cid1 | c: \text{ conf1} , \text{ attSet1} > & \Rightarrow \\
\{\tau\} < oid1 : cid1 | c: \text{ conf2} , \text{ attSet1} > \\
\text{if } & \text{ conf1 } \Rightarrow \{\tau\} \text{ conf2}.
\end{align*}
\]

or rules to lift silent actions among object configurations at the same level of the hierarchy like

\[
\begin{align*}
crl \{\tau\} & \text{ obj1 conf2 } \Rightarrow \text{ obj3 conf2} \\
\text{if } & \{\tau\} \text{ obj1 } \Rightarrow \text{ obj3}.
\end{align*}
\]

are not necessary in the SOS* style.
4 Benchmark

Our benchmark consists of three test cases selected from the literature as archetypal examples of model reconfigurations, transformations and refactorings. In the following we describe the main features of each test case, emphasizing the most relevant details. For the full description of the test cases we refer to the source code of our implementation [12] or to the indicated references.

Architectural reconfiguration. The first test case we consider is the typical reconfiguration scenario in which some components must be migrated from one compromised location to another one. Many instances of this situation arise in practice (e.g. clients or jobs that must be migrated from one server to another one). Some instances of this scenario can be found e.g. in [7, 13]. In what follows we consider a scenario in which components can be nested within each other. Components within an unsafe component $x$ must be migrated into an uncle component $y$ with the additional requirement of changing their status according
to the status of their new container y. Figure 1 depicts one possible instance of the scenario.\(^3\)

The most significant SPO rule\(^4\) is depicted on the left of Fig. 2. It takes an unsafe component and a safe component that are neighbours (they have a common container) and moves the component inside the compromised component one to the safe one while changing its status. More rules are needed (for instance for considering top-level rooms without containers) and some of them have application conditions. As a consequence, the applicability of those rules requires to check the whole model and there is no predefined order on which rules to apply first. The safe system (the system without components in need of evacuation) is reached when no more transformation rules are applicable. For instance, Fig. 1 shows a possible match for the SPO rule. The effect of applying the SPO rule will be to move the normal component under the unsafe one to its new location (under the safe component) while changing its status into safe.

On the right of Fig. 2 instead we the see the main SOS rule: all the components \(c_1\) contained in a unsafe component are evacuated into a safe neighbor component, while changing their status inductively (via \(\text{to}(\text{safe})\)-labelled rules). Figure 1 shows a possible instance of the SOS rule. The effect of the SOS rule will be to migrate the two normal components contained in the unsafe component to the safe component while changing their status (in addition the unsafe component is removed).

**Model translation.** Our second test case is the classical translation of class diagrams into relational database schemes (a description can be found in [14]). The main idea is that classes are transformed into tables and their attributes

---

\(^3\) The figures in the paper follow an intuitive UML-like notation, with boxes for objects and arrows for references. We prefer to use this intuitive notation to sketch the scenarios, leaving the detailed Maude implementation [12] for interested readers.

\(^4\) The big encircled arrow separates the rule's left- and right-hand side. Object identifiers are dropped for the sake of clarity and are to be identified by their spatial location.
into columns of the tables. Associations between classes are transformed into auxiliary tables with foreign keys from the tables corresponding to the associated classes. Figure 3 depicts one possible instance of the scenario.

Figure 4 sketches two illustrative transformation rules. The SPO rule transforms a class (belonging to a package) into a table (within the corresponding schema). It also creates a primary key and the corresponding column for the table. A negative application condition forbids the application of the rule in case the table already exists. The, let us say, corresponding rule in SOS format transforms a class into a table provided that its attributes are transformed into columns and its association ends are properly collected. An auxiliary object is used as a container where to put association ends of the same relation in the same context so that they can be transformed properly by another rule.

Refactoring. The example of model refactoring we consider is the classical attribute pull-up as described in [15]. The main idea is very simple: if all the subclasses of a class c declare the same attribute, then the attribute should be declared at c only. This preserves the semantics of the diagram (as the sub-classes will inherit the attribute) while simplifying it by removing redundancies. Figure 5 depicts one possible instance of the scenario.

Fig. 4. An SPO rule (left) and a SOS rule (right) for model translation

Fig. 5. An instance of the model refactoring test case
Figure 6 depicts two illustrative transformation rules. The SPO rule pulls an attribute up provided that it is not annotated as missing by another class (a set of rules takes care of creating such annotations). The SOS rule instead pulls the attribute up provided that all sibling sub-classes agree to pull it up.

5 Experiments

This section presents our experimental results. Experiments were run on an Ubuntu Linux server equipped with Intel Xeon 2.67GHz processors and 24GB of RAM. Each experiment consists on the transformation of instances of a test case using the discussed representation and transformation styles. Instances are automatically generated with the help of parameterizable instance generators that allow us, for instance, to scale up the instances to check scalability of the various approaches. For each experiment we have recorded the number of rewrites and the running time (not always proportional), put in the y-axis of separate plots. Each experiment is performed for an increased size factor that typically makes the model grow exponentially. The x-axis corresponds to the size of the instance in terms of overall number of objects. The timeout for the experiments is of an hour. We do not present results for instances larger than those where at least one of the techniques already times out (which is denoted by the interruption of the plot).

The goal of the experiments is to collect evidence of performance differences, draw hypotheses about the causes of those differences and validate our hypotheses with further experiments. Our benchmark consists of the three test cases presented in Section 4. The code for replicating our experiments is available at [12].

5.1 SOS vs SOS*

1st experiment. We start testing the impact of our improvement encoding of SOS (SOS*) with a basic set of instance generators. The generator for the reconfiguration test cases has a single parameter which is the depth of the
component containment tree, i.e. for a given natural number \( n \), it generates a binary tree of depth \( n \). The grandfathers of leafs have exactly one unsafe component and one safe component as children. All other components are normal. Figure 1 sketches one such instance. The parameter of the instance generator for the model transformation case is the branching factor of the containment tree, i.e. given for a given natural number \( n \), it generates a UML domain with \( n \) packages, each containing \( 2n \) classes, each containing \( n \) attributes and \( n \) associations. The \( i \)-th association of class \( c \) with \( c \) even (resp. odd) has as opposite the \( i \)-th association of class \( c + 1 \) (resp. \( c - 1 \)). So-built domains have \( n \) packages, \( 2n^2 \) classes and \( n^3 \) association pairs (cf. Fig. 3).

The instance generator for the refactoring test case produces binary trees of class hierarchies. Hence, each class has two sub-classes. In addition each sub-class has one local attribute (that will not be pulled up) and one (non-local) attribute.
inherited from its parent. The topmost class has only one local attribute and one
(non-local) attribute (cf. Fig. 5).

The results of Fig. 7 show a clear superiority of SOS* in most cases. The only
exception is the model translation test case. We argue that there are two reasons.
First, SOS* allows to contextualise some reconfigurations at an arbitrary level of
the nesting hierarchy while SOS has to derive the reconfiguration at the top level
by lifting up silent rewrite steps. The second reason is that SOS* performs less
transformation attempts as it does not try rules that have unnecessary labels.

The reconfiguration test case is a perfect example for both issues. First,
regarding the free contextualisation of top SOS* rules we observe that in the
considered instances the rule can be applied at the bottom of the term, while
the SOS rules require in addition to lift the application of such rule up to the
root. In addition, determining whether a transformation can be carried out
can be determined by the non-applicability of rules in the SOS* case, while in
the SOS case requires to perform many unsuccessful transformation attempts.
In the model translation both styles are essentially equivalent as the top rule
must necessarily apply at the top of the term representing the model and after
transformation the rules are deactivated as the necessary patterns disappear.

2nd experiment. In order to validate the first hypothesis we have performed
experiments where safe components do not accept reconfigurations. In addition
a component whose sub-components are safe becomes safe. This does not only
disable reconfigurations after a migration but also prevents reconfiguration at-
ttempts. The results are depicted in Fig. 8 where it can be seen that now SOS
scales better since the number of rewrite attempts for silent transitions is reduced
(safe components and their containment are not checked for reconfiguration).

3rd experiment. Another improvement of SOS* regards the top-down imposition
of labels in rewrite conditions. In order to validate the effect of top-down enact-
ing of transformations we have conducted further experiments with the model
reconfiguration test case with a different instance generator: now the root is a
normal component, the two sons of the root are an unsafe and a safe component
that contain a fixed number components, each able to change into safe plus
any status of a set of size \( n \), the parameter of the generator. So, for \( n = 0 \), the components to be migrated are able to change into safe, for \( n = 1 \) they are ready to change into safe and another status, and so on. The results of such experiment are depicted in Fig. 9.

As expected the SOS* transformation is not affected as \( n \) increases. Indeed, the SOS* transformation rules will call for a transformation into safe, while in the SOS transformation all possible status changes will be attempted. As a result the computational effort of SOS transformations blows up with the increase of \( n \).

5.2 SPO vs SOS*

In this set of experiments we compare the SPO approach against the SOS* one.

1st experiment. We start with the first set of instance generators used in §5.1.

The results of Fig. 10 show that SOS* is superior in the reconfiguration test case only. The situation can be roughly explained as follows: matching the migration rule consists on finding a subtree whose root is a component having two subtrees: one having an unsafe component as root and one having a safe one as root. In the SPO case the tree is not parsed: indeed we are given a graph and have to check all possible subsets of nodes to see if they constitute indeed a tree. Instead in the SOS case the tree is already parsed (the parsing is a term of the hierarchical representation) which enormously facilitates rule matching (recall that matching amounts to subgraph isomorphism which is NP-complete). As a consequence, the SPO transformation involves more unsuccessful rule attempts and this is the main reason of the drastic difference in running time (and not in number of effective rewrites).

In the rest of the test cases SPO performs better. This is particularly evident in the refactoring test case where the performance of SOS* degenerates mainly due to the lack of a smart transformation strategy. Indeed it can happen that a pull up has to be attempted at some class every time one of the terms corresponding to one of the subclasses changes. Clearly applying rules bottom up would result in better results but this would require a more cumbersome implementation.
We focus now on the transformation test case where we see that SPO performs slightly better. There are various reasons. First, the structure of the model is rather flat. Indeed, the hierarchy is limited to a fixed depth as packages contain classes, classes contain only attributes and associations. So containment trees are of depth 3. In addition, association pairs have to be lifted to the top level in the SOS* transformation since the transformation rule that translates them needs them to be in a common context. This involves an overhead that makes SOS* exhibit a worse performance.

2nd experiment. In order to check the impact of such overhead we have performed an additional experiment in which the instances have no associations at all. Figure 11 shows the results where we see how SOS* is the winner this time confirming our hypothesis.
6 Conclusion

We have presented an empirical evaluation of the performance of two transformation styles that are very popular in rule-based programming and specification. For instance, in the process algebra community they essentially correspond to the rule formats used for specifying reduction and transition label semantics.

We have focused on model transformations and as a result of our experience we have obtained a set of hints that could be useful for future development of model transformations (or other kind of rule-based specifications) in Maude or to enhance the existing ones (e.g. [19]). We think that it might be worth to investigate to which extent our experience can be exported to other rule-based frameworks like CafeOBJ [16], Stratego [17] or XSLT [18] with a particular attention to model transformation frameworks such as MOMENT2-MT [19], ATL [20], Stratego/XT [21], and SiTra [22]. To this aim one should also clarify the influence of Maude’s matching and rewriting strategies in the obtained performances. The study could also be enlarged to other rule styles or alternative implementations of SOS in Maude (e.g. [23, 24]). Another interesting aspect to be investigated is to understand if and how strategy languages (c.f. [23]) or heuristics (c.f. [25]) can be exploited to appropriately guide the model transformation process in the most convenient way.

It is worth to remark that the aim of the paper is not to compare the performance of transformation tools as done in various works and competitions [26, 27]. Rather we assume the point of view of a transformation programmer, which is given a fixed rule-based tool and can only obtain performance gains by adopting the appropriate programming style.

Even if we have focused fundamentally on deterministic transformations many cases (e.g. reconfigurations) are inherently non-deterministic. This gives rise to a state space of possible configurations, whose complexity and required computational effort is clearly influenced by the chosen rule style (evidenced as well by experiments not presented here).
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Interactive Transformations from Object-Oriented Models to Component-Based Models

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Abstract. Consider an object-oriented model with a class diagram, and a set of object sequence diagrams, each representing the design of object interactions for a use case. This article discusses how such an OO design model can be automatically transformed into a component-based model for the purpose of reusability, maintenance, and more importantly, distributed and independent deployment. We present the design and implementation of a tool that transforms an object-oriented model to a component-based model, which are both formally defined in the rCOS method of model driven design of component-based software, in an interactive, stepwise manner. The transformation is designed using QVT Relations and implemented as part of the rCOS CASE tool.

Keywords: Model-driven development, OO design model, sequence diagram, component model, model transformation, QVT

1 Introduction

In the rCOS [3, 12] model-driven design of component-based software, the model of the requirements is represented in a component-based architecture. Each use case is modeled as a component in the requirements model. The interface of the component provides methods through which the actors of the use case interact with the component. The functionality of each method \( m() \) of the interface is specified by pre- and post-conditions \( m() \{ \text{pre} \vdash \text{post} \} \), and the order of the interactions (called the use-case protocol) between the actors and the component as a set of traces of method invocations, graphically represented by a UML sequence diagram. One component may have a required interface through which it uses the provided methods of other components. The linkages (dependency) between components forms a static component-based structure modeled as a component diagram. The types of the variables of the components, i.e. its objects and data, are modeled by a UML class diagram, that has a textual counterpart specification in rCOS. Therefore the model of the component-based architecture of the requirements consists of a model of the component-based static structure (graphically represented as a UML component diagram), a class model (graphically a class diagram), an interaction protocol (graphically a sequence diagram for each component), and a specification of the data functionality of the interface methods.

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In the design, the functionality specification of the interface methods of each component is then refined by decomposition and assignment of responsibilities to objects of the component, obtaining an OO model of object interactions represented by an object sequence diagram. This object sequence diagram refines the sequence diagram of the component (use case). For the purpose of reusability, maintenance, and more importantly, distributed and independent deployment (third party composition) [19], the OO model is abstracted to a model of interactions of components, that is graphically represented as a component sequence diagram defined in the UML profile for rCOS.

This paper presents the design and implementation of a tool for the transformation of a model of object interaction to a model of component interaction. The tool requires user interactions. In each step of interaction, the users decide which objects will be turned into a component, then the tool automatically performs the model transformation. However, we need to define the criteria for the selection of objects to form a component as the validity conditions of the selection. The tool automatically checks the validity, and the transformation of the sequence diagram is carried out if selection passes the check. The transformation also automatically and consistently transform the static structure and reactive behavior (state machine diagram), obtaining a model of component-based design architecture, that correctly refines the component-based architecture of the requirements.

Through a finite number of transformation steps with valid selection on the OO model of each component in the model of requirements, the object sequence diagram is transformed to a component sequence diagram in which the lifelines represent only components. Also, a complete component diagram is generated with the interface protocols as sets of traces and the reactive behavior modeled by state machine diagrams of the components. The transformations of the OO design of all components thus, one by one, obtain a correct refinement of the model of requirements architecture to a component-based design architecture in which each component in the requirements is a composition of a number of components.

The semantic correctness of the transformation and consistency among the different resulting views (diagrams) can be reasoned about within the rCOS framework. The tool is not only applicable in a top-down design process. If object-interaction models can be obtained from packages (modules) of OO programs, the tool can be used to transform OO programs to components, at least on the modeling level. An extension would be required to transform existing source code within a transformation step.

The paper is organized as follows. We start in Section 2 to discuss the concepts of rCOS model to facilitate the definition of the transformation. We present the major principles of the transformation in Section 3, and describe the implementation of the transformation tool. Section 4 shows how the transformation be applied to a case study. Our conclusions and the related work of this paper are discussed in Section 5.

2 UML Profile of rCOS Models

rCOS provides a notation and an integrated semantic theory to support separation of concerns and allows us to factor a system model into models of different viewpoints [3, 12]. The formal semantics and refinement calculus developed based on it are needed for
the development and use of tools for model verification and transformations. The aim of
the development of rCOS tools is to support a component-based software development
process that is driven by automatic model transformations. The model transformations
implement semantic correctness preserving refinement relations between models at dif-
ferent level of abstraction. It is often the case that the models before and/or after a
transformation need to be verified or analyzed, and in that case verification and analysis
tools are invoked. The rCOS project focuses on tool development for model transfor-
mations, and this paper in particular is about the transformation from object-oriented
design models to component-based design models.

UML Profiles [15] are a mechanism to support extending and customizing standard
UML. This mechanism is carried out by defining stereotypes, tagged values and ad-
ditional constraints. Through such a UML profile, rCOS models can be supported by
standard UML infrastructure and CASE tools, minimizing the effort to develop a new
tool, and meeting the requirements for standardization and interoperability.

The rCOS development process involves the following models:

1. The requirements model includes a component diagram, a conceptual class dia-
gram in which classes do not have methods, and a set of sequence diagrams. They
all have their formal rCOS textual counter parts. Also, each method of the provided
interface has a pre- and post-condition specification. The sequence diagrams are
component sequence diagrams in which the lifelines are components, and interac-
tions are inter-component interactions.

2. Each component in the requirements model go through an OO design phase and its
sequence diagram is refined into an object sequence diagram in which each lifeline
is an object, and interactions are intra-object interactions within the component.
The conceptual class diagram in the requirements model is also refined into a de-
sign class diagram in which methods for the intra-object interactions of the object
sequence diagrams are assigned to the classes.

3. Then each OO sequence diagram of a component in the previous stage is abstracted
to a component sequence diagram; thus the component is decomposed into a com-
position of a number of components. After the abstraction transformation is done
for all components of the requirements model, the component-diagram of the re-
quirements is refined to another component diagram with more hierarchical com-
ponents being introduced.

Note that the transformation described here is not limited to rCOS models—rather,
rCOS just prescribes the wellformedness of the input models, and the semantics of the
communication model that will be preserved through the transformation. We refer to
our publications [3,12] for detailed discussions. The rCOS class model is rather a UML
standard class model. In the rest of this section, we define the metamodels of rCOS
components and sequence diagrams.

2.1 The metamodel of rCOS components

The component model is an essential part of rCOS. Its metamodel is defined by a UML
profile diagram shown in Fig. 1, in where an element in the light yellow box represents a
stereotype of rCOS, and the ones in the dark yellow boxes are standard UML metamodel elements. In the metamodel, an rCOS component model consists of:

- **ContractInterface**: Extended from UML Interface, a contract interface provides an interaction point for a component, and defines the static portion of a rCOS interface contract. DesignOperations specify the static functionality of an operation. It is defined as an rCOS design in the form of \( \text{pre} \vdash \text{post} \). An rCOS Field, which is not shown in the figure explicitly, is implemented as a UML Property of a contract interface. (For ease of layout of the diagram, the same ContractInterface element appears twice in Fig. 1.)

- **Protocol**: A contract interface has a Protocol that specifies the traces of invocations to the Operations of the Interface of the contract interface. A protocol contains a StateMachine, a Collaboration and a set of CallEvents. A call event is an invocation of an operation of the contract interface, resulting in the execution of the called operation. Especially, here a Collaboration owns a UML Interaction defined as a RCOSSequenceDiagram, whose metamodel is given in the next subsection.

- **RCOSComponent**: There are two kinds of components in rCOS, ServiceComponents and ProcessComponents. A service component, for short a component here, provides services to the environments through its provided interfaces, and requires services from other components through its required interfaces. rCOS defines separate contracts for the provided interface and required interface of a component. Thus, the metamodel defines one provided contract interface, and optionally a required contract interface.

Fig. 1. The metamodel of rCOS component model
We realize the connection between a component and its provided interface using a UML InterfaceRealization. A UML Usage, a specialized Dependency relationship, is used to link a required interface to its owner component. In addition, we define a stereotype Composition, which is also an extension of UML dependency, to plug a provided interface of a component to a required interface of another component (here, rCOS component operations do not translate naturally to UML component composition). Furthermore, a component may be realized by a set of classes through ComponentRealizations.

2.2 Metamodel of rCOS sequence diagrams

Fig. 2 shows the metamodel of rCOS sequence diagrams. It conforms to the interaction metamodel provided by OMG [15]. In the metamodel, a UML Interaction contains a number of Lifelines, and a set of Messages.

A message specifies a communication from a sender lifeline to a receiver lifeline. It has a sendEvent and receiveEvent which express the MessageOccurrences along the lifelines, appearing in pairs. A message occurrence represents the synchronous invocation of an operation. The BehaviorExecution (green segment of a lifeline in the later diagrams) represents the duration of an operation, and plays no role in our models (yet it is an artefact from the graphical editor).

rCOS has two kinds of sequence diagrams, object sequence diagrams and component sequence diagrams. A lifeline may represent an actor, an object (of a particular class), or a component. When a lifeline represents an object or a component, we call it object lifeline or component lifeline. A CombinedFragment represents a nested block that covers lifelines and their messages to express flow of control, such as an alternative

Fig. 2. Metamodel of rCOS sequence diagram
block (alt) or an iteration block (loop), with their attached boolean guard conditions. Sequence diagrams here do not express recursion.

The two kinds of sequence diagrams are needed to combine both OO design and component-based design in rCOS. The abstract stereotype RCOSSequenceDiagram has subtypes of ObjectSequenceDiagram and ComponentSequenceDiagram, that satisfy the following well-formed conditions, respectively.

1. **ObjectSequenceDiagram:**
   - There is one lifeline representing an *Actor*, and all other lifelines represent objects or components.
   - Messages are *synchronous calls* to an operation provided by the type of the target lifeline, or a constructor/create messages.
   - A message flow starts with a message from the actor to a single component, from components to components or objects, or from objects to objects, but never from objects to components.

   Therefore, and object-sequence diagram can contain both component and object lifelines, and thus also serves as an intermediate data structure for the transformations, until all objects have been transformed.

2. **ComponentSequenceDiagram:**
   - One lifeline represents an *Actor*, and all other lifelines represent components.
   - All receive events occur on the lifelines representing components.
   - Each message is a method call to an operation defined in the provided interfaces of the component represented by the target lifeline.
   - There should be a *composition* relation between two component if there is a message between them in a sequence diagram.
   - No create messages exists in the diagram.

The static semantics, i.e. well-formedness of the rCOS sequence diagrams, including the above conditions, is defined by a set of OCL rules in the rCOS CASE tool. These rules are used to automatically check the well-formedness conditions and the structural consistency of the UML model: for example, the object creation event on a lifeline must precede all other events on the lifeline, and a fragment must include both the sender and the receiver of any event occurring in the fragment.

rCOS also has a dynamic model represented by state diagrams. The metamodel of state diagrams is largely the same as the labelled transition systems provided by standard UML state diagrams, where guarded transitions are again linked to interface methods. We leave the metamodel definition out of this paper.

3 **Transformation from Object- to Component Sequence Diagrams**

We now describe the interactive transformations from an object sequence diagram to a component sequence diagram. The transformations start with an object sequence diagram and a design class diagram. Through a number of steps of interactions between the user and the tool, they generate a component sequence diagram, a component sequence diagram, and the protocols of the provided interface and required interface of each component in the component diagram. In each step, the user selects a set of object
lifelines that she intends to make into a component. The tool will check the validity conditions for this set to form a component. If the selection passes the check, the tool combines the selected object lifelines into a component lifeline, adding a component to the component diagram, and generates the protocols for the component. We describe the principles of the selection and the validity of selection, as well as the generation of a component from the selected lifelines below. As the UML metamodel especially for sequence diagrams is quite verbose as shown in the previous section, we use an alternate, more concise representation here (at the cost of not having established the formal correspondence between the two levels).

3.1 Selection of object lifelines

First, one object lifeline is designated as the controller object of the selection by the user. The principles for picking such a control object not only depend on checkable conditions of the object but also on design considerations of reusability, maintainability, and organization of the system being modeled. The major checkable condition is that this object should be a permanent object in the sequence diagram. This means it should have existed before the start of the execution of the sequence diagram (specified by the precondition of the first message), and it will not be destroyed during the execution (rCOS does not have destructor methods). This also includes software objects representing the control of physical devices, such as barcode readers, controllers of printers, lights, and operating system objects, such as the system clock.

Then the selection of further objects should be made by the user with consideration of the following conditions and principles:

1. any object lifeline that is a receive end of a creation event from a lifeline that is already included in the selection must be selected,
2. the objects in the selection must be strongly connected, i.e. for any lifeline \( \ell \) in the selection there is at least one message path from the controller object to \( \ell \),
3. consider low coupling and high cohesion principle that the selected lifelines have more intensive interaction with each other than with lifelines outside the selection.
4. lifelines that represent objects which will be deployed on different nodes of a distributed system should not be included in the same selection.

The first two conditions are must condition and can be easily checked, as discussed in the next sub-section. The third condition is a desirable principle, and the fourth is a platform dependent condition. The latter two can never lead to an inconsistent model, but to a model that does not capture the intentions correctly, and a detailed discussion of them is out the scope of the paper.

3.2 Validating the lifeline selection

Given an object sequence diagram \( D \), we define some notations for the describing the validation of a selection. We use \( D\text{.lines} \) to denote the set of all lifelines of \( D \), \( D\text{.messages} \) the set of messages, and messages in \( D\text{.messages} \) are represented by \( m[\ell_i, \ell_j] \) for an invocation of \( m \) of \( \ell_j \) from \( \ell_i \). create-messages indicate constructor invocation.
Let $D.selection \subseteq D.lines$ be a selection, and $\ell_c$ the designated controller object, and define $D.rest = D.lines - D.selection$. Further, we define

- $\text{IntraM} = \{m[\ell_i, \ell_j] : \ell_i, \ell_j \in D.selection\}$ \hspace{1cm} \text{Intra-message among the selected lifelines}
- $\text{InM} = \{m[\ell_i, \ell_j] : \ell_i \in D.rest \land \ell_j \in D.selection\}$ \hspace{1cm} \text{Incoming messages to selected lifelines}
- $\text{OutM} = \{m[\ell_i, \ell_j] : \ell_i \in D.selection \land \ell_j \in D.rest\}$ \hspace{1cm} \text{Outgoing messages from selected lifelines}
- $\text{OutsideM} = \{m[\ell_i, \ell_j] : \ell_i, \ell_j \in D.rest\}$ \hspace{1cm} \text{Messages outside the selected lifelines}

A lifeline $\ell$ in sequence diagram $D$ can be either an object lifeline, denoted by $\text{type}(\ell) = \text{Class}$, or a component lifeline, denoted by $\text{type}(\ell) = \text{Component}$. Now we define the conditions below for checking the validity of a selection.

1. All lifelines selected must be object lifelines
   \[\forall \ell \in D.selection \cdot \text{type}(\ell) = \text{Class}\]

2. The controller object $\ell_c$ must be a permanent object. This is done by checking it is not on the receive end of an object creation message.
   \[\forall \ell \in D.lines \cdot (\text{create}[\ell_c, \ell] \notin D.messages)\]

3. The transformation starts with those lifelines that directly interact with the actor, then those directly receiving message from the lifelines that have been made into component lifelines. Therefore any incoming message to the current selection should be from either the actor or a component lifeline
   \[\forall m[\ell_i, \ell_j] \in \text{InM} \cdot (\text{type}(\ell_i) = \text{Actor} \lor \text{type}(\ell_i) = \text{Component})\]

4. Creation messages can only be sent between lifelines inside the selection or between objects outside the selection
   \[\forall \ell_i, \ell_j \in D.lines \cdot (\text{create}[\ell_i, \ell_j] \in \text{IntraM} \lor \text{create}[\ell_i, \ell_j] \in \text{OutsideM})\]

5. Any incoming message to the selection is received either by the controller object or by a lifeline which has a direct path of message from the controller object
   \[\forall m[\ell_i, \ell_j] \in \text{InM} \cdot (\ell_j = \ell_c \lor \exists m[\ell_c, \ell_j] \in \text{IntraM})\]

6. The lifelines of the selection must be strongly connected, meaning that for any selected lifeline $\ell$, there must be a path of messages from the controller object
   \[m[\ell_c, \ell_1], m_1[\ell_1, \ell_2], \ldots, m_k[\ell_i, \ell]\]

Notice that Conditions 4&6 are closure properties required of the section, and that the initial object-sequence diagram of a use case in rCOS always has a use case controller object that satisfies Conditions 2,3&5. Using induction on the number of lifelines, these conditions all together ensures existence of a valid selection for any well-formed sequence diagram that contains object lifelines. Every OO sequence diagram can be translated into the trivial component sequence diagram which internalises all object lifelines into the controller.
3.3 Generating a component from selected lifelines

If the selection passes the validity checking, the transformation will be executed to generate the target models, otherwise an error message is fed back to the tool user. The transformation is specified in the QVT relational notation (see Section 3.4). For the understandability to the formal specification community, we describe the specification in terms of the relation between the source model and the target model, similar to the pre- and postcondition specification of a program.

Given a source sequence diagram $D$, that is an object sequence diagram, and a valid selection $D.selection$, let $D'$ denote the target sequence diagram of one step of the transformation. For a lifeline $\ell$ in $D$ (or $D'$), we use $op(\ell, D)$ (respectively $op(\ell, D')$) to denote the set of method names of the type class of $\ell$ in $D$ (respectively $D'$), $type(\ell, D)$ the type of the lifeline $\ell$ in $D$ (respectively $type(\ell, D')$ in $D'$), and $pIF$ the provided interface of the component that $\ell$ represents if the type of $\ell$ is a component. For a component lifeline $\ell$ in $D$ (or $D'$), $rIF$ denotes the its required interface. We now describe the relation between $D$ and $D'$ as the conjunction of the following predicates.

1. The controller object $\ell_c$ in $D$ is changed to a component lifeline in $D'$

\[
\ell_c \in D.selection \land type(\ell_c, D) = Class \\
\land \ell_c \in D'.lines \land type(\ell_c, D') = Component
\]

2. An incoming message to the selection in $D$ becomes an invocation to the interface methods of $\ell_c$ in $D'$

\[
\forall m[\ell_i, \ell_j] \in InM \cdot (m[\ell_i, \ell_c] \in D'.messages \land m \in pIF(\ell_c))
\]

Notice that the order of the messages and fragments are not to be changed.

3. All the intra-object interactions in the selection in $D$ are collapsed, more precisely hidden inside the component $\ell_c$

\[
\forall m[\ell_i, \ell_j] \in IntraM \cdot (\ell_i, \ell_j \not\in D'.lines \land m[\ell_i, \ell_j] \not\in D'.messages)
\]

4. All the outgoing messages from the selection become outgoing messages from the component that $\ell_c$ represents in $D'$, with the order and fragments preserved, and they become the required methods of the component

\[
\forall m[\ell_i, \ell_j] \in OutM \cdot (m[\ell_c, \ell_i] \in D'.messages \land m \in rIF(\ell_c))
\]

5. No lifelines and messages outside the selection are changed

\[
\forall m[\ell_i, \ell_j] \in OutsideM \cdot (m[\ell_i, \ell_j] \in D'.messages)
\]

From the definition of the resulting sequence diagram $D'$, its static counterparts, the components can be defined. The change for the component diagram can be specified in a similar way. The protocols of the provided interface $pIF(\ell_c)$ and the required interface $rIF(\ell_c)$ of the newly constructed component $\ell_c$ in $D'$ will be generated.

Next, we give an intuition into how the relations defined above can be directly put to use through QVT-Relations.
3.4 Implementation of the transformation

The object sequence diagram to component sequence diagram transformation is implemented through the QVT Relations language using the QVTR-XSLT tool we recently developed [10]. The MOF 2.0 Query/View/Transformation (QVT) [14] is a model transformation standard proposed by OMG. QVT has a hybrid declarative/imperative nature. In its declarative language, called QVT Relations (QVT-R), a transformation is defined as a set of relations between the elements of source metamodels and target metamodels. QVT-R has both textual and graphical notations, and the graphical notation provides a concise, intuitive way to specify transformations.

The QVTR-XSLT tool supports the graphical notation of QVT-R. It provides a graphical editor in which a transformation can be specified using the graphical syntax, and a code generator that automatically generates executable XSLT [21] programs for the transformation. The tool supports in-place transformations so we can focus on defining rules only for the parts of a model we want to change. Multiple input and output models are also supported in the tool.

In the graphical notation, a relation defines how two object diagrams, called domain patterns, relate to each other. Fig. 3 illustrates an example QVT relation in graphical notation which specifies the generation of a component lifeline from an object lifeline. Starting from the root object seq tagged with label ≪Domain≫, the source domain pattern (left part) of the relation consists of a Lifeline lfl with its representing Property under the seq. The target domain pattern (right part) has a similar structure. The patterns are used for structural matching in the source- and target model, respectively.

When the relation is executed, the source domain pattern is searched in the source model. If a match is found, the lifeline and the property are bound to instances of source

![Fig. 3. An example of a QVTR relation](image-url)
model elements. The target domain pattern of the relation acts as a template to create objects and links in the target model. In this example, the target domain pattern creates a *lifeline* object and a *property* object. Both objects own a *name* and an *xmi:id* attributes. These two attributes get values from the corresponding model instances bound by the source domain pattern. Moreover, the *property* object of the target model has now the association *type* set to the component *com*, which is bound (and possible created) by invoking relation *LifelineToCom* in the *when* clause. These clauses specify additional matching conditions and can either refer to other relations, or OCL expressions.

At the implementation level, a complete model consists of a UML model and a DI (diagram interchange) [13] model. The former contains the abstract syntax information that is described in Section 2, and it is stored in Eclipse Modeling Framework (EMF) XMI format, which is supported by many UML CASE tools. The latter contains the layout information in the form of UML 2.0 Diagram Interchange standard [13]. In fact, these two models are technically separate models and saved in different XML files. When the UML model is modified by the transformation, the DI model must be synchronously updated in order to correctly display the corresponding diagrams. The changes to the DI model are also specified using QVT-R, and transformed by the QVTR-XSLT tool. The resulting diagrams for the case study are the result of those transformations after minimal visual cleanup. The transformation is specified by three transformation models. In total, they contain 105 relations, and 45 functions and queries. About 6300 lines of XSLT code are generated for the implementation of the transformation.

To support the rCOS methodology, we have developed a CASE tool [4] with graphical interfaces for designing use cases, classes, component-, sequence- and state diagrams, and the syntactic consistency among these views can be checked. The tool is implemented as an Eclipse-plugin on top of the Eclipse Graphical Modelling Framework and TOPCASED [16]. We have integrated the XSLT programs of the transformation into the user interface of the tool. A user can select a group of lifelines from the interface, and then the XSLT transformation programs are invoked by the tool with these lifelines as parameters. If these lifelines are allowed to become a component, the transformation is executed and the user interface will be automatically refreshed to show the transformation results.

4 Case Study

The Common Component Modelling Example (CoCoME) [3, 17] describes a trading system that is typically used in supermarkets. This case study deals with the various business processes, including processing sales at a cash desk, handling payments, and updating the inventory. The system maintains a catalog of product items, as well as the amount of each item available. It also keeps the historical records of sales; each of them consists of a number of line items, determined by the product item and the amount sold.

At the end of the object-oriented design stage, we get a design model which contains a set of design class diagrams and object sequence diagrams. Fig. 4 shows a simplified version of the design class diagram for the CoCoME example, where the class *CashDesk* is the control class. Fig. 5 depicts the object sequence diagram of use case
process sale, which describes the check out process: a customer takes the products she wants to buy to a cash desk, the cashier records each product item, and finally the customer makes the payment. Applying the transformations discussed in the previous sections, we transform the object sequence diagram into an rCOS component sequence diagram in a stepwise, incremental manner. Meanwhile the object model automatically evolves to a component-based model.
The object sequence diagram of Fig. 5 consists of seven lifelines. The leftmost lifeline is the Actor, and followed by lifelines $L_{\text{CashDesk}}$, $L_{\text{Sale}}$, $L_{\text{LineItem}}$, $L_{\text{Store}}$, $L_{\text{Item}}$ and $L_{\text{Pay}}$, representing objects of class CashDesk, Sale, LineItem, Store, Item and Payment, respectively. Based on our interpretation of the case study, we decide to apply the transformation three times.

The first step deals with the lifeline $L_{\text{CashDesk}}$, which is directly interacting with the actor. Since lifeline $L_{\text{Sale}}$ is created by $L_{\text{CashDesk}}$, and $L_{\text{LineItem}}$ is created by $L_{\text{Sale}}$, they have to be in the same component. As shown in Fig. 6, we select these three lifelines from the sequence diagram, set $L_{\text{CashDesk}}$ as the controller object (main lifeline in the figure), and transform them into a service component $\text{COM}_L_{\text{CashDesk}}$. The component has a provided interface $\text{ConInter}_L_{\text{CashDesk}}$ and a required interface $\text{RInter}_L_{\text{CashDesk}}$. The resulting sequence diagram is shown in Fig. 8, in which lifeline $L_{\text{CashDesk}}$ now represents the new component, and lifelines $L_{\text{Sale}}$ and $L_{\text{LineItem}}$, along with their internal messages, are removed from the diagram.

As we mentioned before, the tool will check whether the selected lifelines can be transformed to a component, and provides an error message if the selection is not valid. For instance, if we choose lifelines $L_{\text{Sale}}$, $L_{\text{Store}}$ and $L_{\text{Item}}$ to become a component, the tool will display an error message, as shown in Fig. 7.

For the second transformation, we select the lifelines $L_{\text{Store}}$ and $L_{\text{Item}}$ from the sequence diagram of Fig. 8, and indicate $L_{\text{Store}}$ as the controller object. Since class Store is composed with class Item, the transformation is allowed, and the two lifelines are transformed into a service component $\text{COM}_L_{\text{Store}}$.

As the result of the second transformation, the lifeline $L_{\text{Store}}$ now represents the component $\text{COM}_L_{\text{Store}}$. Accordingly, the component diagram is changed, where the provided interface $\text{ConInter}_L_{\text{Store}}$ is plugged to the required interface $\text{RInter}_L_{\text{CashDesk}}$ (we only show the final resulting component diagram later in Fig. 11).

For each generated component, we also generate an rCOS protocol, which consists of a sequence diagram and a state diagram, for its provided interface. The protocol for component $\text{COM}_L_{\text{Store}}$ is shown in Figs. 9 & 10. The left part of the sequence diagram in Fig. 9 specifies the interactions of the component with its environment (represented by a fresh actor), and the right part defines the interactions between the com-
ponent and its internal objects. We notice that a message originally sent from a non-selected lifeline and received by another selected lifeline, such as the getPrice message in Fig. 5, now becomes two messages. The first getPrice message is received by the component COM_L_Store, and then delegated to the original receiving lifeline L_Item using the second getPrice message.

In the third transformation, we turn L_Pay, the only object lifeline left, into component COM_L_Pay. Thus we get the final component diagram shown in Fig. 11, which...
Fig. 11. Final component diagram of the CoCoME example depicts the relationships among the three components of the model. We obtain the final component sequence diagram, in which all lifelines represent components, except the one representing the actor (see Fig. 12), fulfilling the structural well-formedness rules of component sequence models as discussed in Section 2.

Through applying the object sequence diagram to component sequence diagram transformation three times, we have successfully developed the design model of CoCoME into a component model. The component model includes component sequence diagrams and component diagrams to define the relationship of components. Each component has its provided/required interfaces, as well as a protocol, that consists of a sequence diagram and a state diagram, to define the behaviors of the component.

Fig. 12. Final rCOS component sequence diagram for usecase process sale
5 Conclusion

A major research objective of the rCOS method is to improve the scalability of semantic correctness preserving refinement between models in model-driven software engineering. The rCOS method promotes the idea that component-based software design is driven by model transformations in the front end, and verification and analysis are integrated through model transformations.

As nearly all existing component-based technologies are realized in object-oriented technologies, most design processes start with an OO development process and then at the end of the process an OO design is directly implemented by using a component-based technology, such .COM or .NET. It is often the case that an OO program is developed first and then it is transformed into component software. Our approach improve this practice by allowing a component-based model of the requirements, and a seamless combination of OO design and component-based design for each components in the requirements. The combination is supported by the interactive transformations from OO design to component-based design presented in this paper, in a stepwise and compositional manner. This allows the object-oriented and component-based design patterns to be used in the OO design and captured in the specification of the transformation.

In the tool implementation, the transformation is specified in a subset of the graphical QVT Relations notation. The correct implementation of the interactive transformation requires the definition of a UML profile of the abstract syntax of the rCOS model that is presented in the paper. The QVT specification of the transformation is automatically transformed to an executable XSLT program, that can be run through an Eclipse-plugin. The presented technique and tool can be combined with reverse engineering techniques for transformation of OO programs into component-based programs.

5.1 Related work

As a natural step of model driven development, object-oriented models are further evolved to component-based models to get the benefits of reusability, maintenance, as well as distributed and independent deployment. Surveys of approaches and techniques for identification reusable components from object-oriented models can be found in [2,20]. Based on the principle of “high cohesion and low coupling”, researchers try to cluster classes into components. The basic ideas are: calculate the strength of semantics dependencies between classes and transform them into the form of weighted directional graph, then cluster the graph using graph clustering or matrix analysis techniques [20]. Using clustering analysis, components with high cohesion and low coupling are expected to be obtained in order to reduce composition cost.

Particularly, since use cases are applied to describe the functionality of the system, the work of [18] focuses on applying various clustering methods to cluster use cases into several components. In [6], the static and dynamic relationships between classes are used for clustering related classes in components, where static relationship measures the relationship strength, and dynamic relationship measures the frequency of message exchange at runtime. COMO [9] proposed a method which measures inter-class relationships in terms of create, retrieve, update and delete (CRUD) operations of model elements. It uses dynamic coupling metric between objects to measure the
potential number of messages exchanged. All above approaches are based on clustering algorithms, which makes them much different from our approach, where transformations are applied at the design stage by a human.

Identifying reusable components from object-oriented models was considered to be one of the most difficult tasks in the software development process [6]. Most existing approaches just provide general guidelines for component identification. They lack more precise criteria and methods [5]. Because of the complexity of source information and the component model itself, it is not advisable for component designers to manually develop component-based models from object-oriented models [20]. Alas, there are almost no (semi)-automatic tools to help designers in the development process [18]. The work of the paper makes a useful attempt to address this problem, and provide a tool supporting.

Sequence diagrams have of course already been used informally in UML-based modeling since their conception. Recently, [7] presents a rigorously defined variant called “Life Sequence Charts” with tool support to use them for system design. The focus there is however not on component modeling, but giving a formal semantics to sequence charts for synthesis.

In [3], we have studied this top-down development process, carried out by hand, for the CoCoME case study. Our process is motivated by an industrial CASE tool, MASTERCRAFT [11]. There, the focus is on the design and refinement of the relational method specifications using the rCOS language [8, 22].

5.2 Future work

There are still many challenges in the automation of model transformations, especially on the level of method specifications, such as applying the expert pattern in the object-oriented design stage. It is not enough to only provide a library of transformations, but more importantly, the tool should provide guiding information on which rule is to be used [12]. Since our methodology (unsurprisingly) coincides with textbook-approaches to design of OO- and component software, we hope that the tool can also become a foundation for education in software engineering. It should guide the user through the different stages with recommendations, e.g. where detail should be added to the model, or where refinement is necessary. Based on metrics, the tool could also propose concrete transformation parameters. It is also difficult to support consistent and correct reuse of existing components when designing a new component. We will continue working in this direction to overcome these challenges.

The rCOS Modeler that implements the transformations discussed here can be downloaded together with examples from http://rcos.iist.unu.edu.

References

Runtime Verification of Temporal Patterns for Dynamic Reconfigurations of Components

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Abstract. Dynamic reconfigurations increase the availability and the reliability of component-based systems by allowing their architectures to evolve at runtime. Recently we have proposed a temporal pattern logic, called FTPL, to characterize the correct reconfigurations of component-based systems under some temporal and architectural constraints. As component-based architectures evolve at runtime, there is a need to check these FTPL constraints on the fly, even if only a partial information is expected. Firstly, given a generic component-based model, we review FTPL from a runtime verification point of view. To this end we introduce a new four-valued logic, called RV-FTPL (Runtime Verification for FTPL), characterizing the “potential” (un)satisfiability of the architectural constraints in addition to the basic FTPL semantics. Potential true and potential false values are chosen whenever an observed behaviour has not yet lead to a violation or satisfiability of the property under consideration. Secondly, we present a prototype developed to check at runtime the satisfiability of RV-FTPL formulas when reconfiguring a Fractal component-based system. The feasibility of a runtime property enforcement is also shown. It consists in supervising on the fly the reconfiguration execution against desired RV-FTPL properties. The main contributions are illustrated on the example of a HTTP server architecture.

1 Introduction

This paper deals with the formal specification and verification of dynamic reconfigurations of component-based systems at runtime. Dynamic reconfigurations increase the availability and the reliability of those systems by allowing their architectures to evolve at runtime.

Dynamic reconfiguration of distributed applications is an active research topic [1,2,21] motivated by practical distributed applications like, e.g., those in Fractal [10] or OSGI³. In many recent works, the idea of using temporal logics to manage applications at runtime has been explored [6,18,8,14].

³ http://www.osgi.org
In [14], we have proposed a temporal pattern logic, called FTPL, to characterize the correct reconfigurations of component-based systems under some temporal and architectural constraints (1). We have also explained in [19], how to reuse a generic formal model to check the component-based model consistency through reconfigurations, and to ensure that dynamic reconfigurations satisfy architectural and integrity constraints, invariants, and also temporal constraints over (re)configuration sequences (2).

As component-based architectures evolve at runtime, there is a need to evaluate the FTPL constraints on the fly, even if only a partial information can be expected. Indeed, an FTPL property often cannot be evaluated to true or false during the system execution. In addition, the reconfigurations change the validity of FTPL constraints by modifying the component architecture. In this paper, given a generic component-based model, we review FTPL from a runtime verification point of view (3). To this end we introduce a new four-valued logic, called RV-FTPL (Runtime Verification for FTPL), characterizing the “potential” (un)satisfiability of the architectural constraints in addition to the basic FTPL semantics. Like in RV-LTL [8], potential true and potential false values are chosen whenever an observed behaviour has not yet lead to a violation or acceptance of the property under consideration.

We then integrate the runtime verification of temporal patterns into the Fractal component model [10]. More precisely, we describe a prototype developed to check at runtime—by reusing the FPath and FScript [12] tool supports—the satisfiability of RV-FTPL formulas. This verification is performed when reconfiguring a component-based system (4). More, the feasibility of a runtime property enforcement is also shown. It consists in supervising at runtime the reconfiguration execution in order to ensure that the RV-FTPL property of interest is fulfilled (5): our 4-valued logic can help in guiding the reconfiguration process.
namely in choosing the next reconfiguration operations to be applied. The main contributions are illustrated on the example of a HTTP server architecture.

The remainder of the paper is organised as follows. After introducing a motivating example in Sect. 2, we briefly recall, in Sects. 3 and 4, the considered architectural (re-)configuration model and the FTPL syntax and semantics. We then define in Sect. 5 the runtime verification of FTPL (RV-FTPL) refining FTPL semantics with potential true and potential false values. Section 6 describes a prototype implementing the RV-FTPL verification, and its integration into the Fractal framework. Section 7 explains how to enforce, at runtime, Fractal component system reconfigurations against desired RV-FTPL properties. Finally, Section 8 concludes before discussing related work.

2 Motivating Example

To motivate and to illustrate our approach, let us consider an example of an HTTP server from [11]. The architecture of this server is displayed in Fig. 2.

The RequestReceiver component reads HTTP requests from the network and transmits them to the RequestHandler component. In order to keep the response time as short as possible, RequestHandler can either use a cache (with the component CacheHandler) or directly transmit the request to the RequestDispatcher component. The number of requests (load) and the percentage of similar requests (deviation) are two parameters defined for the RequestHandler component:

- The CacheHandler component is used only if the number of similar HTTP requests is high.
- The memorySize for the CacheHandler component must depend on the overall load of the server.
- The validityDuration of data in the cache must also depend on the overall load of the server.
- The number of used file servers (like the FileServer1 and FileServer2 components) used by RequestDispatcher depends on the overall load of the server.

![Fig. 2. HTTP Server architecture](image)

We consider that the HTTP server can be reconfigured during the execution by the following reconfiguration operations:
1. AddCacheHandler and RemoveCacheHandler which are respectively used to add and remove the CacheHandler component when the deviation value increased/decreased around 50;

2. AddFileServer and removeFileServer which are respectively used to add and remove the FileServer2 component;

3. MemorySizeUp and MemorySizeDown which are respectively used to increase and to decrease the MemorySize value;

4. DurationValidityUp and DurationValidityDown to respectively increase and decrease the ValidityDuration value.

As an illustration, we specify the AddCacheHandler reconfiguration expressed in the FScript language [12]. When the deviation value exceeds 50, the reconfiguration consists in instantiating a CacheHandler component. Then, the component is integrated into the architecture, and the binding with the required interface of RequestHandler is established. Finally, the component CacheHandler is started.

```
1  action AddCacheHandler (root)
2  newCache = new ("CacheHandler");
3  add ($root, $newCache);
4  bind ($root/child::RequestHandler/interface::getcache, $newCache/interface::cache);
5  start ($newCache);
```

3 Architectural (Re-)Configuration Model

This section recalls the generic model for component-based architectures given in [14] and inspired by the model in [20,21] for Fractal. Both models are graphs allowing one to represent component-based architectures and reconfiguration operations and to reason about them.

Component-based models must provide mechanisms for systems to be dynamically adapted—through their reconfigurations—to their environments during their lifetime. These dynamic reconfigurations may happen because of architectural modifications specified in primitive operations. Notice that reconfigurations are not the only manner to make an architecture evolve. The normal running of different components also changes the architecture by modifying parameter values or stopping components, for instance.

3.1 Component-based architectures

In general, the system configuration is the specific definition of the elements that define or prescribe what a system is composed of. The architectural elements we consider (components, interfaces and parameters) are the core entities of a component-based system and relations over them to express various links between these basic architectural elements. We consider a graph-based representation illustrated by Fig. 3.
In our model, a configuration $c$ is a tuple $\langle \text{Elem}, \text{Rel} \rangle$ where $\text{Elem}$ is a set of architectural elements, and $\text{Rel} \subseteq \text{Elem} \times \text{Elem}$ is a relation over architectural elements.

The architectural elements of $\text{Elem}$ are the core entities of a component-based system:

- **Components** is a non-empty set of the core entities, i.e., components;
- **RequiredInterfaces** and **ProvidedInterfaces** are defined to be subsets of $\text{Interfaces}$. Their union is disjunctive;
- **Parameters** is a set of component parameters.

The architectural relation $\text{Rel}$ then expresses various links between the previously mentioned architectural elements.

- **InterfaceType** is a total function that associates a type with each required and provided interface;
- **Provider** is a total surjective function which gives the component having at least a provided interface, whereas **Requirer** is only a total function;
- **Contingency** is a total function which indicates for each required interface if it is **mandatory** or **optional**;
- **Definer** is a total function which gives the component of a considered parameter;
- **Parent** is a partial function linking sub-components to the corresponding composite component. Composite components have no parameter, and a sub-component must not be a composite including its parent component, and so on;
- **Binding** is a partial function which connects together a provided interface and a required one: a provided interface can be linked to only one required interface, whereas a required interface can be the target of more than one provided interface. Moreover, two linked interfaces do not belong to the same component, but their corresponding instantiated components are sub-components of the same composite component. The considered interfaces must have the same interface type, and they have not yet been involved in a delegation;
Delegate expresses delegation links. It is a partial bijection which associates a provided (resp. required) interface of a sub-component with a provided (resp. required) interface of its parent. Both interfaces must have the same type, and they have not yet been involved in a binding; State is a total function which associates a value from \{started, stopped\} with each instantiated component: a component can be started only if all its mandatory required interfaces are bound or delegated; Last, Value is a total function which gives the current value of a considered parameter.

Complete and formal definitions can be found in [19].

Example 1. Figure 4 gives a graph-based representation of the example from Sect. 2. In this figure, the architectural elements are depicted as boxes and circles, whereas architectural relations are represented by arrows.

3.2 Dynamicity of Component Architectures

To support system evolution, some component models provide mechanisms to dynamically reconfigure the component-based architecture, during their execution. These dynamic reconfigurations are then based on architectural modifications, among the following primitive operations:

- instantiation/destroction of components;
- addition/removal of components;
- binding/unbinding of component interfaces;
- starting/stopping components;
- setting parameter values of components;

or combinations of them. A component architecture may also evolve by modifying parameter values or stopping components, like in the example.

Considering the component-based architecture model recalled in Sect. 3.1, a reconfiguration action is modelled by a graph transformation operation adding or removing nodes and/or arcs in the graph of the configuration. An evolution operation \( \text{op} \) transforms a configuration \( c = \langle \text{Elem}, \text{Rel} \rangle \) into another one \( c' = \langle \text{Elem}', \text{Rel}' \rangle \). It is represented by a transition from \( c \) to \( c' \), noticed \( c \xrightarrow{\text{op}} c' \). Among the evolution operations (running operations and reconfigurations), we particularly focus on the reconfiguration ones, which are either the above-mentioned primitive architectural operations or their compositions. The remaining running operations are all represented by a generic operation, called the \( \text{run} \) operation; it is also the case for sequences of running operations.

The evolution of a component architecture is defined by the transition system \( \langle \mathcal{C}, \mathcal{R}_{\text{run}}, \rightarrow \rangle \) where:
- \( \mathcal{C} = \{ c, c_1, c_2, \ldots \} \) is a set of configurations;
- \( \mathcal{R}_{\text{run}} = \mathcal{R} \cup \{ \text{run} \} \) is a finite set of evolution operations;
- \( \rightarrow \subseteq \mathcal{C} \times \mathcal{R}_{\text{run}} \times \mathcal{C} \) is the reconfiguration relation.

Given the model \( M = \langle \mathcal{C}, \mathcal{R}_{\text{run}}, \rightarrow \rangle \), an evolution path (or a path for short) \( \sigma \) of \( M \) is a (possibly infinite) sequence of configurations \( c_0, c_1, c_2, \ldots \) such that \( \forall i \geq 0. \exists r_i \in \mathcal{R}_{\text{run}}. c_i \xrightarrow{r_i} c_{i+1} \in \rightarrow \).

We use \( \sigma(i) \) to denote the \( i \)-th configuration of a path \( \sigma \). The notation \( \sigma_i \) denotes the suffix path \( \sigma(i), \sigma(i+1), \ldots \) and \( \sigma'_i \) denotes the segment path \( \sigma(i), \sigma(i+1), \sigma(i+2), \ldots, \sigma(j-1), \sigma(j) \). The segment path is infinite in length when the last state of the segment is repeated infinitely often. We write \( \Sigma \) to denote the set of evolution paths, and \( \Sigma^f \) (\( \subseteq \Sigma \)) for the set of finite paths.

**Example 2.** A possible evolution path of the HTTP server is given in Fig. 5. In this path,

- \( c_0 \) is a configuration of the HTTP server without the \text{CacheHandler} nor \text{FileServer2} components;
- \( c_1 \) is obtained from \( c_0 \): the load value was changed following the running of the \text{RequestHandler} component;

![Fig. 5. Part of an evolution path of the HTTP server example](image-url)
– $c'_1$ is the same configuration as $c_1$: Without the CacheHandler component, the RemoveCacheHandler reconfiguration cannot terminate, it is then rollbacked without any modification;
– $c_2$ is obtained from the configuration $c_1$ by adding CacheHandler, following the AddCacheHandler reconfiguration operation;
– $c_3$ is the configuration $c_2$ in which the memorySize value was increased;
– $c'_3$ is the same configuration as $c_3$: The result of the running is not observable;
– $c_4$ is obtained from $c_3$ by adding the FileServer2 component;
– $c_5$ is like the configuration $c_6$ but the durationValidity value was increased.

4 FTPL

In this section, we recall the syntax of the linear temporal logic for dynamic reconfigurations introduced in [14] and called FTPL. It allows characterizing the correct behaviour of reconfiguration-based systems by using architectural invariants and linear temporal logic patterns. FTPL has been inspired by proposals in [15], and their temporal extensions for JML [24,9,17].

Let us first recall the FTPL syntax as presented in [14]. A configuration property, denoted with $conf$, is a first order logic formula over sets and relational operations on the primitive sets and over relations defined in Sect. 3.1. A trace property, denoted with $trace$, is a temporal constraint on (a part of) the execution of the dynamic reconfiguration model. Further, for a reconfiguration operation $ope$, its ending is considered as an event.

| event ::= | ope terminates |
| ope exceptional |
| ope normal |
| trace ::= | always $conf$ |
| eventually $conf$ |
| $trace_1$ A $trace_2$ |
| $trace_1$ V $trace_2$ |
| temp ::= | after event $temp$ |
| before event $trace$ |
| $trace$ until event |

The trace properties specify the constraints to ensure on a sequence of reconfigurations. We mainly specify the always and eventually constraints which respectively describe that a property has to be satisfied by every configuration of the sequence for the former, or by at least one configuration of the sequence for the latter.

Every temporal property concerns a part of the execution trace on which the property should hold: it is specified with special keywords, like e.g., after, before or until a particular event has happened.

The set of FTPL formulae is denoted with $FTPL$. The complete and detailed semantics can be found in [14].

Example 3. Let us now illustrate the FTPL language on the example of the HTTP server from Sect. 2. Notice that the reconfiguration AddCacheHandler (resp. RemoveCacheHandler) adds (resp. removes) CacheHandler when the deviation value is greater (resp. less) than 50:

\[
\text{after RemoveCacheHandler terminates} \\
\text{Property 1 : (eventually deviation}>50 \\
\text{until AddCacheHandler terminates)}
\]

The previous property specifies that the deviation value eventually becomes greater than 50 between the two considered reconfigurations.
5 Runtime Verification for FTPL: RV-FTPL

As component-based architectures evolve at runtime, there is a need to check the FTPL constraints on the fly, even if only a partial information is expected. Indeed, an FTPL property often cannot be evaluated to true or false during the system execution, as only the history of the system is available and no specification of its future evolution exists. In addition, as architectural reconfigurations change the component architecture, they also change the values of FTPL constraints.

In this paper we review the FTPL semantics from a runtime verification point of view. To this end we introduce a new four-valued logic, called RV-FTPL (Runtime Verification for FTPL), characterizing the “potential” (un)satisfiability of the architectural constraints in addition to the basic FTPL semantics. Intuitively, potential true and potential false values are chosen whenever an observed behaviour has not yet lead to a violation or acceptance of the property under consideration.

Let \( S \) be a set and \( R \) a relation over \( S \times S \). \( R \) is a pre-ordering iff it is reflexive and transitive, and a partial ordering iff it is anti-symmetric in addition. For a partial ordering \( R \), the pair \( (S, R) \) is called a partially ordered set; it is sometimes denoted \( S \) when the ordering is clear. A lattice is a partially ordered set \( (S, R) \) where for each \( x, y \in S \), there exists (i) a unique greatest lower bound, and (ii) a unique least upper bound. A lattice is finite iff \( S \) is finite. Every finite lattice has a well-defined unique least element, often called the minimum, and a well-defined greatest element, often called the maximum.

More specifically, let \( B_4 = \{ \bot, \bot, \top, \top \} \) be a set where \( \bot, \top \) stand resp. for false and true values where as \( \bot, \top \) stand resp. for potential false and potential true values. We consider \( B_4 \) together with the truth non-strict ordering relation \( \sqsubseteq \) satisfying \( \bot \sqsubseteq \bot, \bot \sqsubseteq \top, \top \sqsubseteq \top \). On \( B_4 \) we define the unary operation \( \neg \) as \( \neg \bot = \top, \neg \top = \bot, \neg \bot = \top, \neg \top = \bot \), and we define two binary operations \( \land \) and \( \lor \) as the minimum, respectively the maximum, interpreted with respect to \( \sqsubseteq \). Thus, \( (B_4, \sqsubseteq) \) is a finite de Morgan lattice but not a Boolean lattice.

Before defining the RV-FTPL semantics, let us recall that a configuration property \( conf \in FTPL \) is valid on a configuration \( c = \langle \text{Elem}, \text{Rel} \rangle \) when the evaluation of \( conf \) on the configuration \( c = \langle \text{Elem}, \text{Rel} \rangle \) is true, written \( [c \models conf] = \top \); otherwise, the property \( conf \) is not valid on \( c \), written \( [c \models conf] = \bot \).

\textbf{Definition 1 (RV-FTPL Semantics).} Let \( \sigma_n^f \in \Sigma^f \) be a finite execution path of the length \( n + 1 \). Given an FTPL property, its value on \( \sigma_n^f \) is given by the interpretation function \( \llbracket \_ \models \_ \rrbracket_{rv} : \Sigma^f \times \text{FTPL} \to B_4 \) defined as follows:

1. For the configuration properties and events:
\[
\begin{align*}
\sigma_n^0(i) \models \text{conf}\rangle_rv &= \begin{cases} 
\top & \text{if } \sigma_n^0(i) \models \text{conf} = \top \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0(i) \models \text{ope normal}\rangle_rv &= \begin{cases} 
\top & \text{if } 0 < i \leq n \land \sigma_n^0(i-1) \neq \sigma_n^0(i) \\
& \land \sigma_n^0(i-1) \stackrel{\scriptstyle \text{ope}}{\rightarrow} \sigma_n^0(i) \in \rightarrow \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0(i) \models \text{ope exceptional}\rangle_rv &= \begin{cases} 
\top & \text{if } 0 < i \leq n \land \sigma_n^0(i-1) = \sigma_n^0(i) \\
& \land \sigma_n^0(i-1) \stackrel{\scriptstyle \text{ope}}{\rightarrow} \sigma_n^0(i) \in \rightarrow \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0(i) \models \text{ope terminates}\rangle_rv &= \begin{cases} 
\top & \text{if } \text{ope normal} \lor \text{ope exceptional} \\
\bot & \text{otherwise}
\end{cases}
\end{align*}
\]

2. For the trace properties:
\[
\begin{align*}
\sigma_n^0 \models \text{always } \text{conf}\rangle_rv &= \begin{cases} 
\bot & \exists i. (0 \leq i \leq n \land \sigma_n^0(i) \models \text{conf}\rangle_rv = \bot) \\
\top & \text{otherwise}
\end{cases} \\
\sigma_n^0 \models \text{eventually } \text{conf}\rangle_rv &= \begin{cases} 
\top & \exists i. (0 \leq i \leq n \land \sigma_n^0(i) \models \text{conf}\rangle_rv = \top) \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0 \models \text{trace}_1 \land \text{trace}_2 \rangle_rv &= \sigma_n^0 \models \text{trace}_1 \rangle_rv \land \sigma_n^0 \models \text{trace}_2 \rangle_rv \\
\sigma_n^0 \models \text{trace}_1 \lor \text{trace}_2 \rangle_rv &= \sigma_n^0 \models \text{trace}_1 \rangle_rv \lor \sigma_n^0 \models \text{trace}_2 \rangle_rv
\end{align*}
\]

3. For the temporal properties:
\[
\begin{align*}
\sigma_n^0 \models \text{after } \text{temp}\rangle_rv &= \begin{cases} 
\top & \forall i. (0 < i \leq n \land \sigma_n^0(i) \models \text{event}\rangle_rv = \top \Rightarrow \sigma_n^0(i) \models \text{temp}\rangle_rv = \top) \\
\bot & \exists i. (0 \leq i \leq n \land \sigma_n^0(i) \models \text{temp}\rangle_rv = \bot) \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0 \models \text{before } \text{trace}\rangle_rv &= \begin{cases} 
\top & \forall i. (0 < i \leq n \land \sigma_n^0(i) \models \text{event}\rangle_rv = \top \Rightarrow \sigma_n^0(i) \models \text{trace}\rangle_rv \in \{\top, \bot\}) \\
\bot & \text{otherwise}
\end{cases} \\
\sigma_n^0 \models \text{trace until } \text{event}\rangle_rv &= \begin{cases} 
\top & \forall i. (0 < i \leq n \land \sigma_n^0(i) \models \text{event}\rangle_rv = \top \Rightarrow \sigma_n^0(i) \models \text{trace}\rangle_rv \in \{\top, \bot\}) \\
\bot & \text{otherwise}
\end{cases}
\end{align*}
\]

Let us now comment and illustrate the above definition. The goal of our work is to be able to detect when the FTPL properties become false. So, for configuration properties and events, the interpretation does only depend on the fact that considered configurations actually belong to the path \(\sigma_0^0\). For events, the basic FTPL semantics is reflected in the interpretation function.

For trace properties the intuition is as follows.

- The \text{always } \text{conf} property is not satisfied on \(\sigma_0^0\) if there is a configuration of \(\sigma_0^0\) which does not satisfy \text{conf}. For the other cases, the property is evaluated to be "potentially true". Indeed, if the execution terminated in \(\sigma_0^0\), the property would be satisfied.
The eventually conf property is satisfied on $\sigma^n_0$ if at least one configuration of $\sigma^n_0$ satisfies conf. In the other cases, the property is evaluated to be "potentially false". Indeed, if the execution terminated in $\sigma^n_0$, the property would be violated.

Example 4. Figure 6 displays an evolution path of the HTTP example. The next array illustrates the evaluation of two trace properties on each configuration, depending on the chosen either FTPL or RV-FTPL semantics:

\[
\begin{array}{c|c|c|c|c|c|c|c}
\text{configuration} & c_0 & c_1 & c_2 & c_3 & c_4 & c_5 & c_6 \\
\hline
\text{event} & \text{Add} & \text{Remove} & \text{Add} & \text{Remove} & \text{Add} & \text{Remove} & \text{Add} \\
\hline
\text{after} & \text{deviation} & < 50 & < 50 & < 50 & < 50 & < 50 & > 50 \\
\hline
\text{eventually} & \text{deviation} & > 50 & > 50 & > 50 & > 50 & > 50 & > 50 \\
\hline
\text{FTPL} & \top & \top & \top & \top & \top & \bot & \bot \\
\hline
\text{RV-FTPL} & \bot & \bot & \bot & \bot & \bot & \top & \top \\
\end{array}
\]

Fig. 6. Part of an evolution path of the HTTP server example

Considering the FTPL semantics, we cannot conclude about the interpretation of the considered properties, until we reach the configuration $c_5$. On the contrary, in RV-FTPL we say at the beginning that the always property is expected to be true in the future, until we reach $c_5$ where it is false.

The intuition of the definition of temporal properties is as follows:

- The value of the after event temp property is potentially true either if the event event does not occur in all considered configurations, or if the occurrence of the event event on a configuration implies that the temp temporal property is evaluated to true on the suffix of the path starting at this configuration. The after event temp property is evaluated to false if there is a configuration $\sigma^n_0(i)$ of $\sigma^n_0$ where the event event happens and temp is evaluated to false on the suffix $\sigma^n_i$.

- The value of the before event trace property is potentially true if either the event event does not occur in all considered configurations, or if trace is evaluated either to true or to potentially true on the prefix of the path where the event event occurs. The before event trace property is evaluated to false if there is a configuration $\sigma^n_0(i)$ of $\sigma^n_0$ where event happens, and trace is evaluated either to false or to potentially false on the path ending at $\sigma^n_0(i)$, non including this configuration.
– The value of the trace until event property is potentially true if the trace property is evaluated either to true or to potentially true on the prefix of the path where there is a configuration satisfying event, the prefix being without that configuration. The trace until event property is evaluated to false either if there is a configuration $\sigma^n_0(i)$ of $\sigma^n_0$ where event happens, and if trace is either false or potentially false on the path ending at $\sigma^n_0(i)$ but non-including it; or if $\sigma^n_0$ does not satisfy the trace property when event does not happen on $\sigma^n_0$. The property is potentially false if the event event does not occur in all considered configurations.

Example 5. Let us again consider the path in Fig. 6 and the FTPL property 1 after RemoveCacheHandler terminates
(eventually deviation>50
until AddCacheHandler terminates)
explained in Example 3. The following array displays the value of the considered property interpreted respectively in FTPL and in RV-FTPL:

<table>
<thead>
<tr>
<th>Property</th>
<th>FTPL</th>
<th>RV-FTPL</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>? ?</td>
<td>$T^*$</td>
</tr>
<tr>
<td>$c_0$</td>
<td>$T^*$</td>
<td>$T^*$</td>
</tr>
<tr>
<td>$c_1$</td>
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</tr>
<tr>
<td>$c_2$</td>
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<tr>
<td>$c_3$</td>
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<td>$c_4$</td>
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<td>$c_5$</td>
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<tr>
<td>$c_6$</td>
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<tr>
<td>$c_7$</td>
<td>$L^*$</td>
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</tr>
<tr>
<td>$\ldots$</td>
<td>$T^*$</td>
<td>$T^*$</td>
</tr>
</tbody>
</table>

From the FTPL semantics point of view, we cannot conclude about the validity of the property until we reach the configuration $c_6$. Using the RV-FTPL semantics, the property interpretation is potential true before the reconfiguration RemoveCacheHandler is executed. Then, the property value becomes potential false until the deviation becomes greater than 50 on $c_5$; as a consequence the property value becomes potentially true because of partial information.

6 Using RV-FTPL Properties to Check Reconfigurations

The proposals of the paper have been applied to the Fractal component model. This section presents the prototype we have been developing to check at runtime the satisfiability of RV-FTPL formulas on Fractal component-based systems. To this end, it exploits and adapts the FPath and FScript [12] tool supports for Fractal to evaluate the desired RV-FTPL formulas after each reconfiguration operation.

6.1 Overview of Fractal, FPath and FScript

The Fractal model is a hierarchical and reflective component model intended to implement, deploy and manage software systems [10]. A Fractal component is both a design and a runtime entity that consists of a unit of encapsulation, composition and configuration. A component is wrapped in a membrane which can show and control a casually connected representation of its encapsulated
content. This content is either directly an implementation in case of a primitive component, or sub-components for composite components.

In order to control the internal structure of a component at runtime, the Fractal model also defines standard interfaces named controllers. In addition, the Fractal model can be extended thanks to new controllers which allow the user to integrate new features.

FPath [12] is a domain-specific language inspired by the XPath language that provides a notation and introspection mechanisms to navigate inside Fractal architectures. FPath expressions use the properties of components (e.g. the value of a component attribute or the state of a component) or architectural relations between components (e.g. the subcomponents of a composite component) to express queries about Fractal architectures.

FScript [12] is a language that allows the definition of reconfigurations of Fractal architectures. FScript integrates FPath seamlessly in its syntax, FPath queries being used to select the elements to reconfigure. To ensure the reliability of its reconfigurations, FScript considers them as transactions and integrates a back-end that implements this semantics on top of the Fractal model.

6.2 Integrating RV-FTPL Property Verification into Fractal

To check RV-FTPL properties at runtime, we have implemented two Fractal controllers which observe the Fractal component model: our first controller, called the reconfiguration controller, permits capturing reconfiguration invocations, whereas the second controller, called the RV-FTPL controller, handles RV-FTPL formulas.

Figure 7 explains how both controllers are used to evaluate properties of interest. When a reconfiguration is invoked (1), the reconfiguration controller executes the reconfiguration (2)—specified in a FScript file—on the considered component-based architecture. It then invokes the RV-FTPL controller (3) to evaluate the RV-FTPL properties from a file (4) where those properties are specified. The RV-FTPL controller uses the instantiated component model (5) and executes queries over it: to post up the property evaluation result to the user, the RV-FTPL controller parses the property of interest and uses a visitor to evaluate it on the current configuration using FPath. In the case of the future
patterns containing the `after` keyword, the visitor waits for the reconfiguration event before evaluating the temporal part of the property. On the contrary, for the past patterns, i.e., the RV-FTPL properties without the `after` keyword, the trace part of the property is evaluated before the reconfiguration event appears. This avoids us from saving all the previous configurations needed to evaluate the property once the event appears.

![Running prototype](image)

**Fig. 8. Running prototype**

The above verification procedure has been integrated into the EVA4Fractal tool previously described in [13]. Figure 8 shows our prototype in action: a Fractal implementation of the HTTP server example is running and the FTPL property

```
after RemoveCacheHandler terminates
(eventually deviation>50
until AddCacheHandler terminates)
```

is evaluated at runtime after each reconfiguration execution. The reader can notice that after the execution of the reconfiguration `RemoveCacheHandler`, the value of the property is potential false. If the value of deviation raises above 50, when the reconfiguration `AddCacheHandler` is applied, the property value becomes true.

7 Using RV-FTPL Properties to Enforce Reconfigurations

As explained in Sect. 1, one of the main motivations of the present work is to use the RV-FTPL property evaluation to control the execution of reconfigurations.
Actually, for some kind of systems like critical systems or embedded systems, the behaviour where the property evaluation becomes false might be not acceptable. To this end, we can use potential true or potential false values to enforce the reconfigurations.

In this section, we show the capability of our monitor to enforce the component-based system reconfigurations by using the interpretation of desired properties. The principle is illustrated in Fig. 9. While interpreting RV-FTPL properties (1), the potential true or potential false values can be used to guide the choice of the next reconfiguration operation (2) which will be applied to the component architecture (3). Let us give an intuition about our approach:

1. Let us consider the RV-FTPL property 3 valued \( \perp^p \) on the current architectural configuration \( c_5 \) from the path given in Fig. 6;
2. We are looking for enabling the reconfiguration operations that make the component-architecture evolve to a new architectural configuration where the RV-FTPL property will be enforced;
3. The reconfiguration manager chooses the reconfiguration AddCacheHandler to be applied;
4. The property will be enforced: it is valued to \( \top^p \) on the new configuration \( c_6 \).

In Fractal an obvious manner to implement the reconfiguration choice procedure is to reuse the transaction mechanism of FScript [12], allowing the system to rollback to a consistent state when a reconfiguration operation failed. We propose to exploit this mechanism to evaluate the RV-FTPL property on the possible target configurations, until a reconfiguration operation where the system benefits enforcement in the best possible way, is found.

We display in Fig. 10 the execution scenario using this mechanism. For each FScript reconfiguration, a transaction is started and the considered reconfiguration operation is executed. Then, the RV-FTPL property is evaluated on the reached configuration. If the interpretation value is true, there is no need to consider remaining reconfiguration operations, so the transaction is committed and the execution goes on. For other interpretation values, the transaction is rollbacked and the results of the reconfiguration valuation are recorded. When
all the enable reconfigurations are explored, the recorded results are used to choose the most appropriate reconfiguration operation which is then applied to the system. To help this choice, adaptation policies [13,11] defined by the user, or distributed controllers [18] for knowledge-based priority properties, or runtime enforcement monitors [16] built automatically for several enforcable properties, can be used.

If for every reconfiguration operation the property of interest is violated, the execution should be either stopped or continued with special recovery operations, and the user should be informed. This reaction clearly depends on the system features (safety critical systems, embedded systems, etc.). Again, adaptation policies can be used to handle events associated with the property violation on the one hand, and to specify special recovery reconfiguration operations, on the other hand.

8 Conclusion

As component-based architectures evolve at runtime, this paper pays particular attention to checking—on the fly—temporal and architectural constraints expressed with a linear time temporal logic over (re)configuration sequences, FTPL [14]. Unfortunately, an FTPL property often cannot be evaluated to true or false during the system execution. Indeed, only a partial information about the system evolution is available: only a (finite) history of the system state is known, and no specification about its future evolutions exists. To remedy this problem, we have reviewed the FTPL semantics from a runtime verification point of view. Inspired by proposals in [8], we have introduced a new four-valued logic, called RV-FTPL, characterizing the “potential” (un)satisfiability of the architectural FTPL constraints in addition to the basic FTPL semantics.

The paper has also reported on the prototype we have been developing to verify and enforce RV-FTPL properties. Given a Fractal component-based system [10] and some desired temporal and architectural FTPL contraints, to make it possible the system to reconfigure, the prototype interprets RV-FTPL formulas at runtime. The feasability of a runtime property enforcement has also been discussed: the proposed 4-valued logic not only captures information absence,
but also helps the monitor in guiding the reconfiguration process, namely in choosing the next reconfiguration operations to be applied.

**Related work.**

In the context of dynamic reconfigurations, ArchJava [3] gives means to reconfigure Java architectures, and the ArchJava language guarantees communication integrity at runtime. Barringer and al. give a temporal logic based framework to reason about the evolution of systems [5]. In [4], a temporal logic is proposed to specify and verify properties on graph transformation systems.

In the Fractal-based framework, the work in [21] has defined integrity constraints on a graph-based representation of Fractal, to specify the reliability of component-based systems. Unlike [21], our model lays down only general architectural constraints, thus providing an operational semantics to other component-based systems, to their refinements and property preservation issues. On the integrity constraints side, the FTPL logic allows specifying architectural constraints more complex than architectural invariants in [12]. Let us remark that architectural invariants as presented in [12] can be handled within the FTPL framework by using **always** \( cp \), where \( cp \) represents the considered architectural invariant.

Among other applications, our proposals aim at a monitoring of component-based systems. In [6], Basin and al. have shown the feasibility of monitoring temporal safety properties (and, more recently, security properties) using a runtime monitoring approach for metric First-order temporal logic (MFOTL). In [23,22], monitors are used to check some policies at runtime, and to enforce the program to evolve correctly by applying reconfigurations. A similar approach based on a three-valued variant of LTL has been proposed in [7]. Contrary to those works, we focus on temporal and architectural constraints to make it possible component-based systems to reconfigure at runtime.

In [8], a three-valued and a four-valued LTL are studied from a logic point of view. In [16], the authors have studied the class of enforceable properties from the point of view of the well-known temporal property hierarchies. The automatic monitor generation for enforceable properties has also been proposed. In this direction, it would be interesting and important to characterize the FTPL temporal patterns wrt. the class of enforceable properties. For non-enforceable temporal patterns, we intend to exploit event-based adaptation policies to make the system behave and reconfigure according to a given recovery policy when the desired property is violated.

**References**


Timed Conformance Testing for Orchestrated Service Discovery*

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Abstract. Orchestrations are systems deployed on the Internet where there is a central component (called orchestrator) coordinating other components (called Web services), pre-existing to the orchestration design phase. Web services are made available through repositories on the Internet to orchestration designers. Service discovery refers to the activity of identifying Web services offered by third parties. We propose an approach to discover Web services by taking into account the intended behaviors of Web services as they can be inferred from the orchestrator specifications. Web services are tested with respect to those behaviors to decide whether or not they can be selected. Specifications of orchestrators are Timed Input/Output Symbolic Transition Systems. Web service intended behaviors are elicited by means of symbolic execution and projection techniques. Those behaviors can be used as test purposes for our timed symbolic conformance testing algorithm.

Keywords: Web service discovery, orchestrations, conformance testing, timed testing, symbolic execution.

1 Introduction

As explained in [18], the World Wide Web has now evolved from a place where we share and find data to a place where we share and find dedicated functionalities. Such functionalities, called \textit{Web services}, can be assembled to build systems whose particularity is that basic functional units (\textit{i.e.} Web services) are developed and offered by different parties and are physically stored in different places on the Internet. The process of building systems by combining Web services is known as \textit{Web service composition}. Composing Web services may be achieved by means of several architectural approaches. Here we focus on \textit{orchestration architectures} [14,19]. An \textit{orchestration} is a Web service system containing a controller component, called an \textit{orchestrator} which serves as an interface for users and is responsible for coordinating Web services invocations accordingly to the user needs. In order to build orchestrations, the first step is to find required Web

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services: this activity is often referred to as (Web) Service Discovery [20]. Web services must be published and accessible on some known repositories, and must be associated with descriptions allowing the designer to select them. Those descriptions contain usually only functional aspects (what are the offered functionalities), and pieces of information may be syntactic (e.g. what is the interface of a service in terms of offered methods for example). As discussed in [20], such descriptions ground discovery procedures by matching orchestration requirements with descriptions of candidate Web services.

In this paper we aim at completing those existing matching procedures mainly based on static analysis by techniques exploiting Web service executions. Provided that the system designer produces a behavioral description of the orchestrator before the Web service selection phase, we aim at taking benefits of the knowledge of the orchestrator to select Web services. Since the orchestrator is responsible for Web service invocations, orchestrator executions mainly contain sequences of Web service invocations conditioned by Web service reactions. Therefore an orchestrator greatly constrains the set of acceptable behaviors (i.e. sequences of emissions/receptions that are called \textit{traces}) of Web services to be selected. Our proposal is to use that set of acceptable traces to guide a selection procedure based on testing techniques. Technically, orchestrators are specified by means of \textit{Timed Input/Output Symbolic Transition Systems (TIOSTS)}, that we define as an extension of \textit{Input/Output Symbolic Transition Systems} [8,12] to deal with timing issues. Regardless of symbolic representations of data, TIOSTS can be seen as a sub class of Timed Automata [1] with one clock per transition. Taking time into account in our work is mandatory because defining timers and reasoning about them is very common in orchestrator descriptions. Typically, one of the most well known ways to describe orchestrators is the WS-BPEL specification language [9]. Operations that can be made on clocks in TIOSTS reflect the common usage of timers in WS-BPEL, which are used to guard orchestrator reactions, typically in situations when some Web service does not react to stimuli of the orchestrator. [3] provides a systematic and detailed translation of the WS-BPEL language towards a particular family of TIOSTS. Advantages of using TIOSTS are twofold: first, we can take benefits of the formal testing framework that we previously defined [8,10,13] by extending it to timing issues. Secondly, we use symbolic execution techniques to analyze the orchestrator description: from a tree-like structure symbolically representing all possible executions of the orchestrator and by means of projection and mirroring techniques, we transform those behaviors into intended Web service behaviors. From those behaviors we extract test purposes to be used in a testing algorithm. A Web service conforming to the test purpose extracted from the orchestrator becomes a good candidate to be integrated in the orchestration. The testing algorithm is a timed extension of the one we defined in [13] and further adapted to the test of orchestrators in context in [8].

Using formal techniques to evaluate the compatibility of Web services (in particular relatively to timing aspects) in Web service systems has been addressed several times, but very often ([6,7,15,16]) based on verification techniques ap-
plied on (parts of) system models (including Web service models or communication protocol models). On the contrary, we use testing techniques to discover Web services for which no model is supposed to be available (we only have knowledge of their interface to send them inputs and receive their outputs). Our proposal is close to the one given in [2] where the goal is to evaluate conformance of Web services to orchestrators thanks to testing techniques. While they use model-checking algorithms applied to testing without considering timing issues, we use symbolic execution techniques within a timed setting.

2 Timed Input Output Symbolic Transition Systems

2.1 Syntax

TIOSTS are symbolic communicating automata introducing constraints over execution delays of transitions. We represent data by means of classical typed equational logic. A data type signature is a couple \( \Omega = (S, Op) \) where \( S \) is a set of types and \( Op \) is a set of operations, each one provided with a profile \( s_1 \cdots s_{n-1} \rightarrow s_n \) (for \( i \leq n, s_i \in S \)). A set of \( S \)-typed variables is a set \( V \) of the form \( \prod_{s \in S} V_s \). The set of \( \Omega \)-terms with variables in \( V \) is denoted \( T_\Omega(V) = \bigcup_{s \in S} T_\Omega(V_s) \), and is inductively defined as usual over \( Op \) and \( V \). \( T_\Omega(\emptyset) \) is simply denoted \( T_\Omega \). A \( \Omega \)-substitution is a function \( \sigma : V \rightarrow T_\Omega(V) \) preserving types. In the following, we note \( T_\Omega(V) \) the set of all \( \Omega \)-substitutions of the variables \( V \). Any substitution \( \sigma \) may be canonically extended to terms. The identity \( \Omega \)-substitution over the variables of \( V \), \( Id_V \), is defined as \( Id_V(v) = v \) for all \( v \in V \).

The set \( Sen_\Omega(V) \) of all typed equational \( \Omega \)-formulas contains the truth values \( true \), \( false \) and all formulas built using the equality predicates \( t = t' \) for \( t, t' \in T_\Omega(V) \), and the usual connectives \( \neg, \vee, \wedge \). In the sequel, we suppose that a signature \( \Omega = (S, Op) \) is given. \( S \) necessarily contains a distinguished type name \( time \), provided with constant symbols in the so-called set of delays \( D \subseteq \mathbb{R}_+^* \) (the set of strictly positive real numbers) and also provided with: the constant symbols \( 0 \) and \( \infty \) representing the first non countable ordinal, and with the usual arithmetic operators as \( +, -, <, \leq \). Moreover \( D \) is supposed to be stable under addition, i.e. for any \( d, d' \in D \) we have \( d + d' \in D \) and under subtraction, i.e. for any \( d, d' \in D \) with \( d < d' \), then \( d' - d \in D \).

TIOSTS are then defined over so-called TIOSTS signatures. A TIOSTS signature \( \Sigma \) is a tuple \( (\mathcal{V}, C) \), where \( \mathcal{V} \) is a set of data variables, and \( C \) is a set of communication channels. A transition of a TIOSTS is a tuple composed of: a source state, a minimal and a maximal delay for the transition firing, a formula called a guard over variables defining a constraint on variable interpretations for the transition firing, a communication action, an affection on data variables to update variable assignments, and a target state. Communication actions are receptions (inputs, denoted by ?) or emissions (outputs, denoted by !) through channels of \( C \), or the unobservable communication action (denoted \( \tau \)). The set
of communication actions over $\Sigma$ is defined as $\text{Act}(\Sigma) = I(\Sigma) \cup O(\Sigma) \cup \{\tau\}$, where: $I(\Sigma) = \{c?x \mid x \in V, c \in C\}$ and $O(\Sigma) = \{ct | t \in T_\Omega(V), c \in C\}$.

**Definition 1 (TIOSTS).** Let $\Sigma = (V, C)$ be a TIOSTS signature. A TIOSTS over $\Sigma$ is $G = (Q, init, Tr)$ where: $Q$ is a set of state names, $init \in Q$ is the initial state and $Tr \subseteq Q \times (D \cup \{0\}) \times (D \cup \{\infty\}) \times \text{Sen}_\Omega(V) \times \text{Act}(\Sigma) \times T_\Omega(V)^V \times Q$ is a set of transitions.

In the sequel, for any TIOSTS $G = (Q, init, Tr)$ over $\Sigma$, we note $Q_\Sigma$, $\text{init}_G$, $T_{rg}$ and $L_G$ respectively for $Q$, $\text{init}$, $Tr$ and $\text{Act}(\Sigma)$. For any transition $tr \in T_{rg}$ of the form $(q, \delta_{\text{min}}, \delta_{\text{max}}, \psi, \text{act}, \rho, q')$, $\delta_{\text{min}}$ is intuitively the minimum delay to wait before the transition can be fired, and $\delta_{\text{max}}$ is the the maximum delay beyond which the transition can not be fired anymore. If $\delta_{\text{max}}$ is $\infty$, there is no upper delay for the transition firing. The class of constraints that can be expressed concerning time in TIOSTS is a subclass of those that can be expressed in Timed automata: constraints characterizing an interval of possible delays before an action occurrence. Reasoning with that simplified class of constraints simplify the rules characterizing our algorithm in Section 4. We use the notations $\text{source}(tr)$, $\delta_{\text{min}}(tr)$, $\delta_{\text{max}}(tr)$, $\text{guard}(tr)$, $\text{act}(tr)$, $\text{sub}(tr)$, and $\text{target}(tr)$ in order to refer respectively to, $q$, $\delta_{\text{min}}$, $\delta_{\text{max}}$, $\psi$, $\text{act}$, $\rho$, and $q'$.

In the sequel, as in [11, 23], we only consider so-called **strongly responsive** TIOSTS that do not contain an infinite sequence of transitions whose actions are in $O(\Sigma) \cup \{\tau\}$.

![Fig. 1. O: TIOSTS for the Business (B) and Low Cost (L) Hotel Reservation examples.](image_url)

**Example 1.** Figure 1 depicts the Business (B) and Low Cost (L) Hotel Reservation examples. They consist of two simplified versions of an orchestration used to reserve a room in a hotel for some given prices and dates, giving priority to dates while varying the price for the Business version, and giving priority...
to the price while varying the dates for the Low Cost version. Since they are very similar, we abusively represent them in the same figure. The only difference consists in transitions labeled with $\alpha$ and $\beta$, for which we provide the guards, communication actions and affectations in the table. Data variables $V$ are $\{\mathtt{dates}, \mathtt{price}, \mathtt{conf}, \mathtt{rstat}, \mathtt{rdates}, \mathtt{rprice}\}$. Communication channels $C$ are $\{u, w\}$ ($u$ to communicate with the user, and $w$ with the Hotel Web service). In both cases, the orchestrator ($O$) receives the desired dates and price from the user (transition $q_0 \to q_1$) and tries to find a room by using the Hotel Web service (transition $q_1 \to q_2$). The answer from the Hotel Web service must arrive before 60 seconds, else a timeout error message is sent to the user (transition $q_2 \to q_0$). This answer can be: (1) ‘reserved’, if a room was found and reserved for those dates and price, (2) ‘option’, if a room was found with a date and/or price close to the ones given as input, and (3) ‘noRooms’, indicating that there are no available rooms at all. According to the answer from the Hotel Web service, the orchestrator may in turn: (1) confirm the reservation; (2) notify the user that it is not possible to find a room: it may be due to the answer ‘noRoom’ from the Hotel Web service or it is ‘option’ and, (2.1) for the Business version (see $\alpha$ in the figure for $w \to B$), the dates of the optional reservation are not the ones given by the user; (2.2) for the Low Cost version (see $\alpha$ in the figure $w \to L$), the price of the optional reservation is not the one given by the user; (3) if the answer is ‘option’ and: (3.1) for the Business version (see $\beta$ in the figure for $w \to B$), the dates are the ones desired by the user but the price is different (usually higher), then the user is asked to confirm the new price before trying again to make the reservation (transitions $q_3 \to q_4 \to q_1$); (3.2) for the Low Cost version (see $\beta$ in the figure for $w \to L$), the price is the one desired by the user but the dates are different, then the user is asked to confirm the new dates (transitions $q_3 \to q_4 \to q_1$).

### 2.2 Semantics

In order to associate semantics to TIJOSTS, we begin by interpreting data occurring in $\Omega$: an $\Omega$-model is a set $M$ whose elements are associated with a type in $S$, and we note $M_s \subseteq M$ the subset of $M$ whose elements are associated with $s$. Moreover for each $op : s_1 \cdots \times s_{n-1} \to s_n$ in $Op$, $M$ is associated with a function $\overline{op} : M_{s_1} \times \cdots \times M_{s_{n-1}} \to M_{s_n}$. We define $\Omega$-interpretations as applications $\nu$ from $V$ to $M$ preserving types and extended to terms in $T_{P}(V)$. $M^\nu$ is the set of all $\Omega$-interpretations of $V$ in $M$. A model $M$ satisfies a formula $\varphi$, denoted by $M \models \varphi$, if and only if, for all interpretations $\nu$, $M \models_\nu \varphi$, where $M \models_\nu t = t'$ is defined by $\nu(t) = \nu(t')$, and where the truth values and the connectives are handled as usual. Given a model $M$ and a formula $\varphi$, $\varphi$ is said satisfiable in $M$ if there exists an interpretation $\nu$ such that $M \models_\nu \varphi$. In the sequel, we sup-

---

3 For concision purpose, several inputs (resp. outputs) can be grouped together in a single transition. Such a feature is practical to model orchestrators and does not raise technical difficulties in our framework, where they can be seen as inputs or outputs of structured pieces of data.
pose that an $\Omega$-model $M$ is given such that all operations of the time sort are interpreted as expected.

Then we associate TIOTS with automaton where messages and delays between them are interpreted in $M$. Such automaton are called Timed Input Output Labeled Transition Systems (TIOLTS) [5,17,21] and are simply automata whose transitions are labeled either by actions (inputs, outputs, or the internal action $\tau$) or by delays.

**Definition 2 (TIOLTS).** Let $L = (L_i, L_o)$ such that $L_i \cap L_o = \emptyset$ and $(L_i \cup L_o) \cap (\{\tau\} \cup \mathbb{R}_+^*) = \emptyset$. A TIOLTS over $L$ is a tuple $G = (Q, \text{init}, \text{Tr})$ where $Q$ is a set of states, $\text{init} \in Q$ is the initial state and $\text{Tr} \subseteq Q \times (L \cup \{\tau\} \cup D) \times Q$ is a set of transitions.

Elements of $L_i$ and $L_o$ are actions that are respectively called inputs and outputs. In the sequel, we will often assimilate $L$ with $L_i \cup L_o$: for example, $l \in L$ will mean $l \in L_i \cup L_o$ and so on. Only transitions carrying elements of $D$ represent delays; other transitions are instantaneously triggered. For any $\text{tr} = (q, a, q')$ of $\text{Tr}$, source($\text{tr}$), act($\text{tr}$) and target($\text{tr}$) stand respectively for $q$, $a$ and $q'$.

For any TIOTS signature $\Sigma$, Elements of $I(\Sigma)$ and $O(\Sigma)$ can be interpreted as actions in the sense of Definition 2: for any $\nu \in M^V$, we note $\nu(c?x)$ for $c?\nu(x)$, and $\nu(\text{act})$ for $c_!\nu(t)$. We note $L_T^G = \{\nu(i) | i \in I(\Sigma) \land \nu \in M^V\}$, $L_o^G = \{\nu(a) | a \in O(\Sigma) \land \nu \in M^V\}$ and $L^G = (L_T^G, L_o^G)$.

Any TIOTS over $\Sigma$ can then be associated with a TIOLTS over $L^G$, by buildings TIOLTS-transitions reflecting all possible firing of all symbolic transitions: roughly, for any transition $\text{tr}$ we identify all the possible couples delay/interpreted action and for each of them, we build two consecutive transitions, the first one labeled by the delay and the second one labeled by the action.

**Definition 3 (Runs of transitions).** For any TIOTS $G$, let $Q_T^G$ stand for $Q_T \times D \times M^V$. For any $\text{tr} \in T_G$, $\text{Run}(\text{tr}) \subseteq (Q_T^G \times D \times Q_T^G), (Q_T^G \times (L^G \cup \{\tau\}) \times Q_T^G)$, is such that $((q_i, t_i, \nu^i), d, (q_i, t_d, \nu^{i_d})), ((q_f, t_d, \nu^f), l, (q_f, t_d, \nu^{f_d})) \in \text{Run}(\text{tr})$ if and only if $q_i = \text{source}(\text{tr})$, $q_f = \text{target}(\text{tr})$, $\nu^i \models \text{guard}(\text{tr})$, $t_d = t_i + d$, $\delta_{min}(\text{tr}) \leq d \leq \delta_{max}(\text{tr})$ and:

if $\text{act}(\text{tr})$ is of the form $c!t$ (resp. $\tau$), then $\nu^f = \nu^i \circ \text{sub}(\text{tr})$, and $l = \nu^i(c!t)$ (resp. $\tau$);

if $\text{act}(\text{tr})$ is of the form $c?x$, then there exists $\nu^a$ such that $\nu^a(z) = \nu^i(z)$ for every $z \neq x$, $\nu^f = \nu^a \circ \text{sub}(\text{tr})$, and $l = \nu^a(c?x)$.

The set of transitions of the TIOLTS associated to a TIOTS contains all those occurring in all runs of all TIOTS transitions. Moreover we add initialization transitions and transitions denoting that whenever no reactions (delays or outputs) are specified from a given state the time may elapse. That TIOLTS is defined as follows.

**Definition 4 (TIOTS unfolding).** The unfolding of $G$ is the TIOLTS $G = (\{\text{init}\} \cup Q_T^G, \text{init}, \text{Tr})$ over $L^G$, where $\text{init}$ is a(n arbitrary) state satisfying
Initialization transitions: for any ν ∈ Mν, (init, τ, (initG, 0, ν)) ∈ Tr,
Run transitions: for any tr ∈ TrG, for any (Q1, d, Q2). (Q2, l, Q3) ∈ Run(tr),
we have (Q1, d, Q3) ∈ Tr and (Q2, l, Q3) ∈ Tr,
Time elapsing transitions: for any Q ∈ {init}∪QG s.t. for all tr ∈ Tr with
source(tr) = Q, act(tr) ∈ LΣ∗, then for any d ∈ D we have (Q, d, Q) ∈ Tr.

The semantics of G is the set of all sequences of actions and delays that
can be associated to G. Such sequences, called timed trace, are defined from the
set of paths of G, denoted Path(G) ⊆ TrG∗ containing the empty sequence ε and all sequences tr1...trn such that source(tr1) = initG, and for all i < n, target(tri) = source(tri+1). Let p be a path of G, the trace of p is the sequence tr(p) = ε if p = ε, and tr(p) = tr(p').act(t) (resp. tr(p) = tr(p')) if p is of the form p'.t and act(t) ̸= τ (resp. act(t) = τ). Traces(G) is the set of traces of all paths
of Path(G). For a trace g of the form g'.d.g'' and for a decomposition d = d1+d2,
the trace g'.d1.d2.g'' is called a decomposition of g. The decomposition operation
can be reiterated for all delays occurring in the trace. Similarly, the reverse
operation, called the composition operation, consists in transforming the trace
g'.d1.d2.g'' into the trace g.d.g'. The set of all traces that can be obtained by
applying both decomposition and composition operations on g as many times as
desired is denoted Time(g), and more generally, for a set S of traces, we note
Time(S) = ∪g∈S Time(g). We note TTraces(G) = Time(Traces(G)).

We define the semantics of G as Sem(G) = TTraces(G).

3 Web Service Discovery: Testing Framework

Regarding the question of testing Web services from timed behaviors of orchestrators, we now present our technical results using preferentially TIO LTS than
TIOSTS. Indeed it is commonly accepted that implementations are modeled as
TIO LTS since black-box testing induces an observational point of view that leads
the tester to perceive the implementation directly as a set of traces. Moreover
any TIOSTS can be associated with its unfolding (see Definition 4).

3.1 Timed conformance relation

Implementations are considered to be TIO LTS which accept any input at any
moment [22] and in such a way that time is correctly modeled from an observational
point of view (most importantly, time is elapsing if no message occurs).

Definition 5 (Implementation). An implementation over L is a strongly re-
sponsive TIO LTS (Q, init, Tr) over L satisfying the following properties:

\[ A^* \] denotes the set of words where letters are in A, ε denotes the empty word and
w1.w2 represents the concatenation of the words w1 and w2.
Input enableness: \( \forall q \in Q, \forall a \in L_i, \exists q' \in Q \text{ such that } (q, a, q') \in Tr \)

Time additivity: \( \forall q_1, q_2, q_3 \in Q, \forall d_1, d_2 \in D, ((q_1, d_1, q_2) \in Tr \land (q_2, d_2, q_3) \in Tr) \Rightarrow (q_1, d_1 + d_2, q_3) \in Tr \)

Time decomposition: \( \forall q_1, q_2, q_3 \in Q, \forall d_1, d_2 \in D, ((q_1, d_1, q_2) \in Tr \Rightarrow \exists q \in Q, ((q_1, d_1, q) \in Tr \land (q, d_2, q_3) \in Tr) \Rightarrow (q_1, d_1 + d_2, q_3) \in Tr \)

\( \tau \) closure: \( \forall q_1, q_2, q_3 \in Q, \forall d_1, d_2 \in D, ((a_1, a_2) = (\tau, d) \lor (a_1, a_2) = (d, \tau)) \Rightarrow ((q_1, a_1, q_2) \in Tr \land (q_2, a_2, q_3) \in Tr) \Rightarrow (q_1, d, q_3) \in Tr \)

Time elapsing: \( \forall q \in Q, \exists (a, q') \in (L_o \cup \{\tau\} \cup D) \times Q \text{ such that } (q, a, q') \in Tr \)

Property 1. Let \( \mathbb{I} \) be an implementation over \( L \). Then \( \text{Traces}(\mathbb{I}) = TT\text{races}(\mathbb{I}) \).

The well-known so-called ioco conformance relation ([22]) defined for IOLTS without time has already been extended to take time into account. Our definition is similar\(^5\) to the ones of [4, 5, 17, 21] which basically include any time delays in the set of observable outputs.

Definition 6 (tioco). Let \( \mathcal{G} \) be a TIOLTS over \( L \) and let \( \mathbb{I} \) be an implementation over \( L \). \( \mathbb{I} \) conforms to \( \mathcal{G} \), denoted \( \text{tioco } \mathcal{G}, \mathbb{I} \), if and only if:

\[ \forall \mathcal{G} \in \text{Traces}(\mathcal{G}), \forall a \in D \cup L_o, \exists a \in \text{Traces}(\mathbb{I}) \Rightarrow \mathcal{G} \in \text{Traces}(\mathcal{G}) \]

Other variants of timed conformance relations have been proposed (see [21] for a detailed presentation).

3.2 Testing Web service from orchestrator behaviours

We first introduce technical operations (projection, mirror and composition) that we will perform on orchestrators to elicit expected behaviors for Web services.

Definition 7 (Projection). Let \( \mathcal{G} = (Q, \text{init}, Tr) \) be a TIOLTS over \( L \). Let \( L' = (L'_i, L'_o) \) with \( L'_i \cap L'_o = \emptyset, L'_i \subseteq L \) and \( L'_o \subseteq L \). The projection of \( \mathcal{G} \) on \( L' \) is the TIOLTS over \( L' \) defined as \( \mathcal{G}_{\parallel L'} = (Q, \text{init}, Tr_{\parallel L'}) \) where \( Tr_{\parallel L'} = \{(q, a, q')_{\parallel L'} \mid (q, a, q') \in Tr \} \) with \( (q, a, q')_{\parallel L'} = (q, a, q') \) if \( a \in L' \cup D \), and \( (q, a, q')_{\parallel L'} = (q, \tau, q') \) otherwise.

In particular, by only considering labels of \( L' \), we consider that transitions carrying labels of \( L \setminus L' \) are performed, but are no more observable: this explains why these labels are simply translated as \( \tau \) in \( \mathcal{G}_{\parallel L'} \). This operation corresponds to the hiding operation [10,23] encapsulating some designated pieces of interface. The projection \( \parallel L' \) can canonically be extended to paths and traces. The mirror operation changes the status (input or output) of actions: it simply depends on the construction of \( L' = (L'_i, L'_o) \).

Definition 8 (Mirror). Let \( \mathcal{G} \) be a TIOLTS over \( L \). The Mirror of \( \mathcal{G} \), \( M(\mathcal{G}) \), is the TIOLTS \( \mathcal{G} \) over \( L' = M(\mathcal{G}) = M(L) \), with \( M(L) = (L'_i, L'_o) \).

---

\(^5\) The slight technical differences are essentially inherited from symbolic executions involved in the test case generation algorithm.
The mirror application can be applied to all elements (transitions, paths, traces) issued from TIOLTS by simply exchanging the role of input and output actions. The mirror operation is often used to design a system interacting with a targeted system: test cases are typically such reactive systems which may be defined by using the mirror operation on the reference model \( \mathcal{G} \). Roughly speaking, as test cases send messages expected by \( \mathcal{G} \) and wait for emissions specified in \( \mathcal{G} \), inputs and outputs are reversed both in the traces of \( \mathcal{G} \) and the test case, until a verdict is computed.

Systems can be composed by taking into account communications between them. As usual, input and output actions will be synchronized when they share the same name. The passing of time will also be synchronized by requiring that any global elapsed time result from the synchronization of subsystem transitions carrying the same delay value. Thus, the global system shares its components exactly the same perception of time. This means that our system composition correspond to locally deployed component-based systems. Particularly in the case of an orchestrator communicating with Web services, this means that the modeling of a Web service is composed of a remote Web service and of message transmissions on the Internet: all the sending and receptions of messages will be stamped with the same clock than the orchestrator clock.

**Definition 9 (Composition).** Let \( \mathcal{G}_1 \) and \( \mathcal{G}_2 \) be two TIOLTS respectively over \( L^{\mathcal{G}_1} \) and \( L^{\mathcal{G}_2} \) such that \( L^{\mathcal{G}_1} \cap L^{\mathcal{G}_2} = L^{\mathcal{G}_1} \cap L^{\mathcal{G}_2} = \emptyset \).

\( \mathcal{G}_1 \otimes \mathcal{G}_2 \) is the TIOLTS \((Q, \text{init}, \text{Tr})\) over \( L^{\mathcal{G}_1} \otimes L^{\mathcal{G}_2} \) with \( L^{\mathcal{G}_1 \otimes \mathcal{G}_2} = L^{\mathcal{G}_1} \cup L^{\mathcal{G}_2} \) and \( L_{\mathcal{G}_1 \otimes \mathcal{G}_2} = (L_{\mathcal{G}_1} \cup L_{\mathcal{G}_2}) \setminus (L_{\mathcal{G}_1} \cap L_{\mathcal{G}_2}) \), \( Q = Q_{\mathcal{G}_1} \times Q_{\mathcal{G}_2} \) and \( \text{init} = (\text{init}_{\mathcal{G}_1}, \text{init}_{\mathcal{G}_2}) \). \( \text{Tr} \) is defined as follows:

**Handshake:** if \( (q_1, a, q'_1) \in \text{Tr}_{\mathcal{G}_1} \) and \( (q_2, a, q'_2) \in \text{Tr}_{\mathcal{G}_2} \) with \( a \in L^{\mathcal{G}_1} \otimes L^{\mathcal{G}_2} \), then \( ((q_1, q_2), a, (q'_1, q'_2)) \in \text{Tr} \).

**Asynchronous execution:** for any \( (q_1, a, q'_1) \in \text{Tr}_{\mathcal{G}_1} \) where \( a \notin L^{\mathcal{G}_2} \cup D \), then for any \( q_3 \in Q_{\mathcal{G}_2} \) we have \( ((q_1, q_3), a, (q'_1, q'_3)) \in \text{Tr} \) (a similar definition holds by reversing the roles of \( \mathcal{G}_1 \) and \( \mathcal{G}_2 \)).

**Property 2.** For \( p \in \text{Path}(\mathcal{G}_1 \otimes \mathcal{G}_2) \) with \( q = \text{tr}(p) \), for \( i \in \{1, 2\} \) we define \( p_{\mathcal{G}_i} \) and \( \varrho_{\mathcal{G}_i} \), as paths and traces over \( L^{\mathcal{G}_i} \):

- if \( p = \varepsilon \) then \( p_{\mathcal{G}_1} = \varepsilon \) and \( \varrho_{\mathcal{G}_1} = \varepsilon \),
- if \( p = p' \cdot ((q_1, q_2), a, (q'_1, q'_2)) \) with \( \varrho' = \text{tr}(p') \) where the last transition is a handshake transition, then \( p_{\mathcal{G}_1} = p'_{\mathcal{G}_1} \cdot (q_1, a, q'_1) \) and \( \varrho_{\mathcal{G}_1} = \varrho'_{\mathcal{G}_1} \cdot a \).
- if \( p = p' \cdot ((q_1, q_2), a, (q'_1, q'_2)) \) with \( \varrho' = \text{tr}(p') \) where the last transition is an asynchronous execution with \( a \notin L^{\mathcal{G}_2} \cup D \), then \( p_{\mathcal{G}_1} = p'_{\mathcal{G}_1} \cdot (q_1, a, q_1) \), \( \varrho_{\mathcal{G}_1} = \varrho'_{\mathcal{G}_1} \cdot a \), \( p_{\mathcal{G}_2} = p'_{\mathcal{G}_2} \) and \( \varrho_{\mathcal{G}_2} = \varrho'_{\mathcal{G}_2} \). A symmetric reasoning holds for an asynchronous transition with \( a \notin L^{\mathcal{G}_1} \cup D \).

Then by construction, for \( i = 1, 2 \), \( p_{\mathcal{G}_i} \in \text{Path}(\mathcal{G}_i) \) and \( \varrho_{\mathcal{G}_i} \in \text{Traces}(\mathcal{G}_i) \). Moreover, we have \( \varrho_{\mathcal{G}_i} = \text{tr}(p)_{\mathcal{G}_i} = \text{tr}(p)_{\mathcal{G}_i} \cdot \text{tr}(p)_{1\mathcal{L}_{\mathcal{G}_i}} \cdot \text{tr}(p)_{1\mathcal{L}_{\mathcal{G}_i}} \).

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\( ^6 \) Let us remark that labels of \( L^{\mathcal{G}_1} \cap L^{\mathcal{G}_2} \) change of status between \( \mathcal{G}_1 \otimes \mathcal{G}_2 \) and \( p_{\mathcal{G}_1} \) and \( p_{\mathcal{G}_1} \) and \( p_{\mathcal{G}_1} \).
Property 3. Let $L_{1,2} = (L_{1}^{i} \cap L_{2}^{o}, L_{1}^{o} \cap L_{2}^{i})$ and $L_{2,1} = M(L_{1,2})$

Let $p_{1} \in \text{Path}(G_{1})$ and $p_{2} \in \text{Path}(G_{2})$ such that $tr(p_{1} \downarrow L_{1,2}) = tr(p_{2} \downarrow L_{2,1})$.

Then there exists a path $p$ of $G_{1} \otimes G_{2}$ such that $pG_{1} = p_{1}$ and $pG_{2} = p_{2}$.

An orchestrator interacts on one hand with the end-user and on the other hand with Web services. Orchestrator actions will be split accordingly\(^7\).

**Definition 10 (Orchestrator).** An orchestrator is a TIOLTS $O = (Q, \text{init}, Tr)$ over $L$ with a distinguished set $L^{W} = (L_{i}^{W}, L_{o}^{W})$ with $L_{i}^{W} \subseteq L_{i}$ and $L_{o}^{W} \subseteq L_{o}$ of so called Web service actions and satisfying the so-called “consistent Web service invocation” property:

There do not exist $a \in L_{i}^{W} \cup D$ and a trace $\varrho$ on $L^{W}$ issued from two distinct paths $p_{1}, p_{2}$ of $O$, that is $tr(p_{1})_{\downarrow L^{W}} = tr(p_{2})_{\downarrow L^{W}} = \varrho$, such that:

- there exists a path of $O$ of the form $p_{1}, p_{1}'$ with $tr(p_{1}, p_{1}')_{\downarrow L^{W}} = \varrho.a$
- for all paths of $O$ of the form $p_{2}, p_{2}'$, we have $tr(p_{2}, p_{2}')_{\downarrow L^{W}} \neq \varrho.a$.

In the sequel, we also call Orchestrator a TIOTS $O$ over $(V, C)$ with a distinguished set $C^{W} \subseteq C$ such that if we note $O$ the unfolding of $O$ and $L^{W}$ the set of all numeric actions built on $C^{W}$ then $O$ is an orchestrator.

The “consistent Web service invocation” property simply expresses that if there are two distinct contexts (paths $p_{1}$ and $p_{2}$) from the orchestrator point of view that are perceived as similar from the Web service ($p_{1}$ and $p_{2}$ have a common projected trace $\varrho$ on $L^{W}$), then the orchestrator should anticipate exactly the same set of reactions from the Web service. It means that designing an orchestrator should take into account that the set of possible Web service reactions depends only on the observational context as perceived by the Web service: only traces projected on $L^{W}$ are relevant to define observational contexts.

**Definition 11 (Require\(_{O}(W)\)).** Let $O$ be an orchestrator over $L$ with a distinguished subset $L^{W}$. Let $W$ be an implementation over $L^{W} = M(L^{W})$.

$W$ satisfies requirements issued from $O$, noted as Require\(_{O}(W)$, if:

For all paths $p$ of $O \otimes W$ such that the set $\{tr(p_{O}).a \mid a \in L_{i}^{W} \cup D, p_{O} \downarrow (\varrho, a, q') \in \text{Path}(O)\}$ is not empty, there exists at least a path $p, p'$ of $O \otimes W$ such that $tr(p, p') = tr(p).a'$ with $a' \in L_{i}^{W} \cup D$.

The property Require\(_{O}(W)$ means that at any reachable state (target state of $p$) in the resulting system $O \otimes W$, $W$ meets the expectations of $O$ if $W$ provides at least one of the behaviors specified by $O$ at this point: the behaviors are either a possible input coming from $W$ or a delay synchronizing behaviors of $O$ and $W$. In other words, to satisfy the Require\(_{O}(W)$ property, $W$ cannot cause a deadlock\(^8\) in the system $O \otimes W$: the path $p, p'$ is precisely an extension in $O \otimes W$ of the path $p$ according to a reaction ($a' \in L_{i}^{W} \cup D$) of $W$ expected by $O$.

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\(^7\) For simplicity purpose, we will consider that orchestrators interact only with one Web service.

\(^8\) In practice, $W$ can be specified over any set $L'$ containing at least $M(L^{W})$.

\(^9\) Another interesting but stronger condition would consist in requiring that the Web service should be able to provide all $a$ in $L_{i}^{W} \cup D$ that extend paths in $O$. 

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Theorem 1. Let $O$ be an orchestrator over $L$ with $L^W \subseteq L$ and let $W$ be an implementation over $M(L^W)$.

$$W \text{ tioco } M(O \downarrow L^W) \Rightarrow \text{Require}_O(W)$$

By Theorem 1, in order to know whether or not a Web service implementation $W$ is suitable to be integrated with a given orchestration (i.e. satisfies the $\text{Require}_O(W)$ property), it suffices to test it accordingly to the $\text{tioco}$ conformance relation and with respect to the behavior deductible from $O$ model using the mirror and projection operations along $L^W$.

Proof. Let us suppose that $W \text{ tioco } M(O \downarrow L^W)$ and let us show that $\text{Require}_O(W)$ holds. Let us consider a path $p$ of $O \otimes W$ such that there exists a path $p_\mathcal{O}.(q,a,q')$ of $O$ with $\text{tr}(p_\mathcal{O}.(q,a,q')) = \text{tr}(p_\mathcal{O}).a$ with $a \in L^W \cup D$.

It exists $a' \in L^W_o \cup D$ and $p'$ such that $p_{\mathcal{W}}.p'$ is a path of $W$ whose trace is $\text{tr}(p_{\mathcal{W}}).a'$. Indeed, the time elapsing property allows to extend paths with transitions carrying actions in $L^W_o \cup \{\tau\} \cup D$. If the considered action would be $\tau$, then we we can reapply the property until getting an action different from $\tau$. $W$ is strongly responsive: it cannot contain an infinite sequence of $\tau$ action.

As $W$ conforms to $M(O \downarrow L^W)$, since $\text{tr}(p_{\mathcal{W}})$ is a trace of both $M(O \downarrow L^W)$ and $W$, this means that $\text{tr}(p_{\mathcal{W}}).a'$ with $a' \in L^W_o \cup D$ is also a trace of $M(O \downarrow L^W)$. Thanks to the “consistent Web service invocation” property, $p_\mathcal{O}$ can be extended as a path $p_\mathcal{O}.p''$ with $\text{tr}(p_\mathcal{O}.p'') = \text{tr}(p_\mathcal{O}).a'$.

Paths $p'$ of $W$ and $p''$ of $O$ share the same projection on $L^W$, the common part of $L^O$ and $L^W$. By Prop.3, they can be synchronized in $O \otimes W$: there exists a path $p.p'$ s.t. $\rho_{\mathcal{W}} = p'$, $\rho_{\mathcal{O}} = p''$ and $\text{tr}(p.p) = \text{tr}(p_\mathcal{O}).a'$ with $a' \in L^W_o \cup D$.

## 4 Symbolic Timed Testing

Following our previous works [8, 13], our testing algorithm is based on symbolic execution techniques.

### 4.1 Symbolic Execution

Symbolically executing a TIOSTS comes to represent its possible executions as a tree structure. Any path of the tree represents in a symbolic way a set of traces associated to a path of the TIOSTS. In the sequel we consider that a set $F = \bigcup_{s \in \mathcal{S}} F_s$ (disjoint of any set of variables introduced in TIOSTS signatures) is given. We also consider a set $F_D$ of time variables (typed on $D$). We note $\text{Sen}(F_D)$ the set of all conjunctions of formulas $x \leq d$ or $d' \leq x$ with $x \in F_D$, $d \in D \cup \{\infty\}$ and $d' \in D \cup \{0\}$. In order to store pieces of information concerning the possible traces of a path we use symbolic states. Those informations are: the last state of the path, the symbolic values assigned to variables, and the constraints on those symbolic values as well as on delays between communication actions occurring in the path. A symbolic state $\eta$ is a tuple $(q, \sigma, \pi, \vartheta)$, where $q \in Q$, $\sigma \in T_D(F)^V$, $\pi \in \text{Sen}_D(F)$, and $\vartheta \in \text{Sen}(F_D)$. In the sequel we note $S$
the set of all such symbolic states. For any $\eta \in \mathcal{S}$ of the form $(q, \sigma, \pi, \vartheta)$, we use the notations $\text{state}(\eta)$, $\text{sub}(\eta)$, $\pi(\eta)$ and $\vartheta(\eta)$ to refer respectively to $q$, $\sigma$, $\pi$, and $\vartheta$. $\eta$ is said satisfiable if and only if there exists $\nu \in M^F$ such that $\nu \models \pi$ and $\nu' \models \vartheta$. $S_{\text{sat}}$ is the set of all satisfiable symbolic states.

**Definition 12 (Symbolic execution of a transition).** Let $\mathcal{G}$ be a TIOSTS over $\Sigma = (V, C)$. Let $\Sigma_F$ stand for $(F, C)$. For any $\tau \in \mathcal{T}_{\mathcal{G}}$ and $\eta \in \mathcal{S}$, such that $\text{source}(\tau) = \text{state}(\eta)$, a symbolic execution of $\tau$ from $\eta$ is a triple $(\eta, \text{sd} \cdot \text{sa}, \eta') \in \mathcal{S} \times (F_D \cdot \text{Act}(\Sigma_F)) \times \mathcal{S}$, such that $\text{sd}$ is a fresh variable, $\text{state}(\eta') = \text{target}(\tau)$, $\vartheta(\eta') = \vartheta(\eta) \land (\delta_{\text{max}}(\tau) \leq \text{sd}) \land (\text{sd} \leq \delta_{\text{max}}(\tau))$, and:

- if $\text{act}(\tau) = c! t$ (resp. $\tau$) then $\text{sa} = c! z$ (resp. $\text{sa} = \tau$) where $z$ is a fresh variable, $\text{sub}(\eta') = \text{sub}(\eta) \circ \text{sub}(\tau)$, and $\pi(\eta') = \pi(\eta) \land \text{sub}(\eta)(\text{guard}(\tau)) \land z = \text{sub}(\eta)(t)$,
- if $\text{act}(\tau) = c! x$ then there exists $\sigma \in T_D(F)^V$ satisfying $y \neq x \Rightarrow \sigma(y) = \text{sub}(\eta)(y)$, and $\sigma(x)$ is a fresh variable such that $\text{sa} = c!\sigma(x)$, $\text{sub}(\eta') = \sigma \circ \text{sub}(\tau)$, and $\pi(\eta') = \pi(\eta) \land \text{sub}(\eta)(\text{guard}(\tau))$.

In the following, $\text{str}$ denotes a triple $(\eta, \text{sd} \cdot \text{sa}, \eta')$ and notations $\text{source}(\text{str})$, $\text{act}(\text{str})$, and $\text{target}(\text{str})$ refer to, respectively, $\eta$, $\text{sd} \cdot \text{sa}$, and $\eta'$.

**Definition 13 (Symbolic execution of a TIOSTS).** A symbolic execution of $\mathcal{G}$, denoted $\text{SE}(\mathcal{G})$, is a couple $\langle \text{init}, \mathcal{R}_{\text{sat}} \rangle$, where $\text{init} = \langle \text{init}_\mathcal{G}, \sigma_0, \text{true}, \text{true} \rangle$ is a symbolic state such that $\forall x \in V$, $\sigma_0(x) \in F$ and $\sigma_0$ is injective, and $\mathcal{R}_{\text{sat}} \subseteq \mathcal{S}_{\text{sat}} \times (F_D \cdot \text{Act}(\Sigma_F)) \times \mathcal{S}_{\text{sat}}$ is the restriction to $\mathcal{S}_{\text{sat}}$ of the relation $\mathcal{R} \subseteq \mathcal{S} \times (F_D \cdot \text{Act}(\Sigma_F)) \times \mathcal{S}$ such that for all $\eta \in \mathcal{S}$ and $\tau \in \mathcal{T}$ with $\text{source}(\tau) = \text{state}(\eta)$, there exists exactly one symbolic execution of $\tau$ from $\eta$ in $\mathcal{R}$. Moreover, for any $(\eta_1, \text{sd}_1, c \Delta z, \eta'_1)$ and $(\eta_2, \text{sd}_2, d \Delta w, \eta'_2)$ in $\mathcal{R}$ with $\Delta \in \{!, ?\}$, we have $\text{sd}_1 \neq \text{sd}_2$ and $z \neq w$.

Note that the symbolic execution is unique, up to the choice of the involved fresh variables.

The symbolic execution of a TIOSTS can be associated with a set of traces that is exactly the one associated to the TIOLOTS denoting its unfolding. Traces of a path $p = (\text{init}, \text{sd}_1, s_{a_1}, \eta_1) \cdots (\eta_{n-1}, \text{sd}_n, s_{a_n}, \eta_n)$ are the traces of the form $\nu'(\text{sd}_1.).\nu(s_{a_1}) \cdots \nu'(\text{sd}_n.).\nu(s_{a_n})$ with\(^{10}\) $\nu \in M^F$ and $\nu' \in D^{F_D}$ two interpretations such that $\nu \models \pi(\eta_0)$ and $\nu' \models \vartheta(\eta_n)$.

### 4.2 Algorithm

In order to assess tioco-conformance of an implementation $\ll$, the key point is that delays appearing in $\ll$ may be formulated differently than they appear in traces of $\text{SE}(\mathcal{G})$: the way they are observed depends on the periodicity of observation in the testing architecture. Therefore our algorithm has to compare traces of $\ll$ to timed traces of $\text{SE}(\mathcal{G})$ defined up to delay composition and decomposition.

\(^{10}\) We apply the convention that $\nu(\tau)$ is the empty word.
Test Purpose: Our algorithm will take behaviors to be tested as inputs in order to pilot the testing process. Such behaviors are called test purposes. Those behaviors are characterized as finite paths of so-called symbolic execution trees which are couples \( ST = (init, \mathcal{R}) \) such that \( init \in S_{sat} \) and \( \mathcal{R} \subseteq S_{sat} \times F_d.\text{Act}(\Sigma_F) \times S_{sat} \). Typical examples of symbolic execution trees are symbolic execution trees of TIOSTS but we will use other structures in Section 5 for Web service elicitation. Test purposes are finite subtrees of \( ST \) whose last transition of each path is not labeled by an input.

Definition 14. Let \( ST = (init, \mathcal{R}) \) be a symbolic execution tree. A \( ST \)-test purpose is \( TP = (init, \mathcal{R}_{TP}) \) where \( \mathcal{R}_{TP} \subseteq \mathcal{R} \) is a finite set s.t. for any \( str \in \mathcal{R}_{TP} \) then either \( \text{source}(str) = init \) or there exists \( str_1 \cdots str_j \) for some \( j \geq 1 \) s.t.:

- for all \( i \leq j \), \( str_i \in \mathcal{R}_{TP} \), \( \text{source}(str_1) = init \) and \( \text{target}(str_j) = \text{source}(str) \),
- for all \( i \leq j - 1 \), \( \text{target}(str_i) = \text{source}(str_{i+1}) \),
- if there is no \( str' \in \mathcal{R}_{TP} \) s.t. \( \text{source}(str') = \text{target}(str) \) then \( \text{act}(str) \notin F_D.I(\Sigma_F) \).

We introduce some technical notations related to test purpose:

a) \( \text{Accept}(TP) \subseteq S_{sat} \) is the set of all \( \eta \) satisfying:
\[
(\exists str \in \mathcal{R}_{TP}, \eta = \text{target}(str)) \land (\forall str \in \mathcal{R}_{TP}, \eta \neq \text{source}(str)).
\]

b) The set \( \text{Reach}(\eta, TP) \) is the set of all symbolic states reachable from \( \eta \) in \( TP \). It contains \( \eta \) and all \( \eta' \) such that there exists a sequence \((\eta, \text{act}_1, \eta_1)(\eta_1, \text{act}_2, \eta_2) \cdots (\eta_{n-1}, \text{act}_n, \eta')\) of transitions of \( \mathcal{R}_{TP} \).

c) We note \( \text{Accept}(\eta, TP) = \text{Reach}(\eta, TP) \cap \text{Accept}(TP) \) the set of all states of \( \text{Accept}(TP) \) which are reachable from \( \eta \).

d) \( \text{targetCond}(\eta) \) is the condition \( \bigvee_{\eta' \in \text{Accept}(\eta, TP)} (\pi(\eta') \land \vartheta(\eta')) \)

Sometimes we write \( \eta \in TP \) to signify that \( \eta \) occurs in some transition of \( \mathcal{R}_{TP} \).

Rule based algorithm: Before giving the rules of our algorithm, we introduce the notion of context \((\eta, f_d, f_s, \theta)\). While interacting with \( I \), we build testing traces. We have to identify paths of \( ST \) that admit them as traces. A context denotes the target state of such a path. Moreover it also contains pieces of information to identify symbolic values with concrete ones (those occurring in the testing trace). It is composed of a symbolic state \( \eta \in S_{sat} \) and of two formulae: \( f_d \) expresses constraints induced by the sequence of data exchanged with \( I \) while \( f_s \) expresses constraints on delays. Finally, in order to identify when the observation occurred, we introduce a duration \( \theta \in D \). The meaning of the context can be intuitively understood as follow:
The trace observed until now can be seen as a trace of the form $q \theta$, where $q$ is a trace of the path leading to $\eta$. Interpretations of variables that occur in communication actions of the path have to satisfy $f_a$ in order to be consistent with values observed in $q$. In the same way interpretations of symbolic delays of the path have to satisfy $f_t$ to be consistent with concretely observed delays in $q$. $\eta$ may have been reached $\theta$ units of time ago.

As there may be many contexts compatible with a testing trace, we use sets of contexts generically noted $SC$ (for Set of Contexts). Sequences of stimuli and observations built by the interaction between the algorithm and I are modeled as elements of $Traces[I]$. Practically, an observation, noted $obs(r)$, is given by $r \in D \cup L^\omega$. A stimulus, noted $stim(i)$, is given by $i \in L^\omega$.

We define several technical notations to denote evolutions of set of contexts:

a) $NextTrigger(a, SC)$, where $a \in L^\Sigma \cup \{\tau\}$, is the set of all contexts that can be reached by triggering a transition of $ST$ consistently with the action $a$.

$(\eta', f'_d, f'_i, \theta') \in NextTrigger(a, SC)$ if and only if $\theta' = 0$ and, if $a$ is of the form $C \triangle \Sigma$ with $\triangle \in \{?, !\}$ (resp. $\tau$) then there exists $(\eta, f_d, f_i, \theta) \in SC$, and $(\eta, \_d, \_i, \eta') \in R$ (resp. $(\eta, \_d, \_i, \eta') \in R$), s.t. $f'_a = f_a$ (resp. $f_d$), $f'_t = f_t \land \theta = s_d$, and both $f'_d \land \pi(\eta')$ and $f'_i \land \theta(\eta')$ are satisfiable.

b) $Wait(d, SC)$, where $d \in D$, is the set of contexts obtained by waiting while progressing to a situation where a transition can be triggered.

$(\eta', f'_d, f'_i, \theta') \in Wait(d, SC)$ if there exists $d' > d$, $a \in L^\Sigma \cup \{\tau\}$ and $C = (\eta, f_a, f_i, \theta) \in SC$ s.t. $NextTrigger(a, \{(\eta, f_d, f_i, \theta + d')\}) \neq \emptyset$, $\eta' = \eta$, $f'_d = f_d$, $f'_t = f_t$, and $\theta' = \theta + d$.

c) $\Delta(d, SC)$, where $d \in D$, is the set of contexts obtained by observing a quiescence situation.

$(\eta', f'_d, f'_i, \theta') \in \Delta(d, SC)$ if and only if there exists $(\eta, f_a, f_i, \theta) \in SC$ such that $\eta' = \eta$, $f'_t = f_t$, $\theta' = \theta + d$ and if we note $\text{react}(\eta)$ the set of all transitions of $R$ whose action is not of the form $sd.i$ with $i \in I(\Sigma)$, $\delta(\eta)$ the formula reduced to true if $\text{react}(\eta) = \emptyset$ and equal to $\land_{i \in \text{react}(\eta)} \neg \pi(\text{target}(str))$ otherwise, then $f'_d$ satisifies $f_d \land \delta(\eta)$ and $f'_i$ satisifies.

d) $TimeElapsed(d, SC) = Wait(d, SC) \cup \Delta(d, SC)$ represents the set of contexts reachable from $SC$ after having waited $d$ time units.

For any set of contexts $SC$, we note:

$Skip(SC) = \{(\eta, f_a, f_i, \theta)(\eta, f_d, f_i, \theta) \in SC, \eta \in TP, \text{(targetCond}(\eta) \land f_d \land f_t) \text{ is satisfiable}\}$,

$Pass(SC) = \{(\eta, f_a, f_i, \theta) \in Skip(SC), \eta \in Accept(TP)\}$.

We use $Skip$ and $Pass$ for shortcuts to $Skip(SC)$ and $Pass(SC)$ when the context is clear. Each verdict is described by means of inference rules holding on sets of contexts. Those rules are of the form: $\text{cond}(ev)$, where $SC$ is a set of contexts, $Result$ is either a set of contexts or a verdict, and $cond(ev)$ is a set
of conditions including the observation \( obs(ev) \) or the stimulus \( stim(ev) \). Such rules express that, given the current set of contexts \( SC \), if \( cond(ev) \) is verified then the algorithm may achieve a step with \( ev \) as elementary action.

As in [13], our algorithm provides four verdicts: \( FAIL \), when the behavior belongs neither to \( TP \) nor to \( ST \) (Rule 3); \( INCONC \), (for inconclusive) when the behavior belongs to \( ST \) and not to \( TP \) (Rule 2), \( PASS \) when the behavior belongs to a path of \( TP \) ending by an accept state and not to any other path of \( ST \) (Rule 5); and \( WeakPASS \), when the behavior belongs to a path of \( TP \) ending by an accept state and to at least one other path of \( ST \) (Rule 4).

**Rule 0:** Initialization

\[
\{(init, true, true, 0)\}
\]

**Rule 1:** No observed outputs for a delay \( d \), consistently with reaching an accept state.

\[
SC \xrightarrow{TimeElaps(d, SC)} obs(d), \text{ Skip} \neq \emptyset, \text{ Pass} = \emptyset
\]

**Rule 1 (bis):** Observation of an output \( o \), consistently with reaching an accept state.

\[
SC \xrightarrow{NextTrigger(o, SC)} obs(o), \text{ Skip} \neq \emptyset, \text{ Pass} = \emptyset
\]

**Rule 2:** The set of reached contexts is outside the test purpose.

\[
SC \xrightarrow{INCONC} SC \neq \emptyset, \text{ Skip} = \emptyset
\]

**Rule 3:** The set of reached contexts is empty.

\[
SC \xrightarrow{FAIL} SC = \emptyset
\]

**Rule 4:** One accept state is reached but not all reached states are accept ones.

\[
SC \xrightarrow{WeakPASS} Pass \neq \emptyset, SC \neq Pass
\]

**Rule 5:** All reached states are accept ones.

\[
SC \xrightarrow{PASS} Pass \neq \emptyset, SC = Pass
\]

**Rule 6:** The tester stimulates by sending an input \( i \).

\[
SC \xrightarrow{NextTrigger(i, SC)} stim(i), SC \neq \emptyset, \text{ Skip}(NextTrigger(i, SC)) \neq \emptyset
\]

Regarding to our contribution in [8, 13], a real novelty here is Rule 1 which computes the impact of time passing on the set of current contexts with respect to the test purpose. Note that rule 6 is the only one that can be applied non deterministically with respect to others. An application strategy defines a test case generation algorithm. The goal of the generic algorithm is similar to the one of the algorithm described in [4]. The two main differences are that we handle data symbolically and our test purposes are defined as symbolic trees instead of properties given as automata. As we will automatically elicit intended behaviors for Web services as symbolic tree from orchestrator specifications, our testing approach is suitable for service discovery.
5 Elicitation of Web service test purposes

For any (TIOSTS) orchestrator \( O \) we identify the interface corresponding to the Web service to be elicited: it is a subset \( C^W \) of the set of channels \( C \) of \( O \). We note \( SE(O) = (init, R_{sat}) \). We note \( (init, R^P) \) the couple reflecting the projection on channels of \( C^W \) in \( SE(O) \). Formally, for any \( str \in R_{sat} \), if we note \( act(str) \) as \( \delta.a \) then we have: if \( a \) is of the form \( c?u \) with \( c \in C^W \) and \( \Delta \in \{?,!\} \) then \( str \in R^P \), else \( (source(str), \delta, \tau, target(str)) \in R^P \). We then apply a mirror operation: we consider the couple \( (init, R^W) \) such that for any \( str \in R^P \), if we note \( act(str) \) as \( \delta.a \) we have: if \( a \) is \( \tau \) then \( str \in R^W \); if \( a \) is of the form \( c?u \) then \( (source(str), \delta, c?u, target(str)) \in R^W \); if \( a \) is of the form \( c!u \) then \( (source(str), \delta, c!u, target(str)) \in R^W \). \( (init, R^W) \) forms a symbolic tree that is a symbolic counterpart to the TIOPTS \( M(\bar{O}_{\bar{L}W}) \) (where \( \bar{O} \) is the unfolding of \( O \)), and we use it to extract test purposes for testing some candidate Web service \( W \) in order to evaluate the validity of \( Require_C(W) \) thanks to Theorem 1.

Example 2. Figure 2 shows the elicited behaviors for the Hotel Web service participating in both versions of the Hotel Reservation example (Business \( B \) and Low Cost \( L \)).

The only difference between the elicited behaviors from the Business and Low Cost versions is in the path condition \( \pi_2 \), where for the former version the dates are kept unchanged, while for the other one the price is kept unchanged.

Let us suppose that we want to test an implementation of a Hotel Web service that is to be used in the Business version. Even if a given implementation of the Web service could be used in both versions, we want to find an implementation of the Web service that does not modify the dates. Moreover, we would expect that a room is found in the first iteration or at least in the second one. Thus, we define the symbolic state \( \eta_3 \) as the only accept state, and if we reach it we also check the path condition \( \pi_2 \) (either the answer is ‘reserved’ or it is ‘option’ with the dates kept unchanged) in order to determine if the Web service behaves as expected. Then, we can know if it fulfills the expectations of the orchestrator. Even if the Hotel Web service answers with a different date, or if no room is found, no \textit{FAIL} verdict would be emitted. However, those are not the behaviors that we expect. Thus, this example shows that, in order to use a Web service within an orchestration so that it can precisely provide behaviors expected by the orchestrator, it has to be tested against test purposes covering these targeted behaviors. Obviously, the choice of behaviors that should be primarily ensured by Web services to be integrated depends on the subjective analysis of the orchestration designer. This methodological subjectivity is similar to the one guiding the choice of appropriate test purposes in a testing activity.

6 Conclusion

In this paper we have shown how to elicit from an orchestrator specifications, intended behaviors of Web services likely to interact with it, and we have shown

\[\text{In the figure, we only show the information related to } \eta_3 \text{ since it is the accept state.}\]
how to use them as test purposes at the Web service discovery phase. Orchestrator specifications are given in a symbolic way and include timing constraints. We have identified a property reflecting the absence of deadlock in an orchestration by relating the orchestrator and Web services of the orchestration. This property serves as reference to select candidate Web service and we have defined a theorem grounding an approach to test this property. Technically our testing approach comes to test the conformance of Web services to symbolic behaviors obtained by symbolically executing the orchestrator specification and by applying projection and mirroring techniques.

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Realizability of Choreographies for Services Interacting Asynchronously

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Abstract. Choreography specification languages describe from a global point of view interactions among a set of services in a system to be designed. Given a choreography specification, the goal is to obtain a distributed implementation of the choreography as a system of communicating peers. These peers can be given as input (e.g., obtained using discovery techniques) or automatically generated by projection from the choreography. Checking whether some set of peers implements a choreography specification is called realizability. This check is in general undecidable if asynchronous communication is considered, that is, services interact through message buffers. In this paper, we consider conversation protocols as a choreography specification language, and leverage a recent decidability result to check automatically the realizability of these specifications by a set of peers under an asynchronous communication model with a priori unbounded buffers.

1 Introduction

Specification and analysis of interactions among distributed components play an important role in service-oriented applications. A choreography is a specification of interactions, from a global point of view, among a set of services participating in a composite service to be designed. One important problem in choreography analysis is figuring out whether a choreography specification can be implemented by a set of distributed peers which communicate using message passing. Even if these peers are obtained by projection [17, 22] from the choreography specification, this does not ensure that they precisely implement the corresponding choreography. This problem is known as realizability.

Most of the work dedicated to realizability assumes a synchronous communication model, see for instance [22, 7, 6, 20]. Only a few works focused on the study of this problem considering an asynchronous communication model, that is communication using message queues or buffers. Fu et al. [11] proposed three conditions that guarantee a realizable conversation protocol. Bultan and Fu [5] also recently defined some sufficient conditions to test realizability of choreographies specified with collaboration diagrams. But defining such conditions is quite restrictive because if they are not satisfied, nothing can be concluded about the system (choreography and peers) being analysed. In [23], the authors refine and extend this former work with an automatic check for bounded asynchronous
communication. The realizability of bounded MSC graphs has also been studied and some decidability results presented in [3]. Su et al. state in [24] that “it remains an open problem whether the realizability problem is decidable”.

More recently, [9] proved the quasi-static scheduling problem of scheduling a set of non-deterministic communicating processes so as to ensure boundedness of buffers, to be undecidable in general, and identified a decidable subclass.

In this paper, we consider conversation protocols [11, 12] (CPs) as choreography specification language, and propose an approach to check automatically the realizability of these specifications by a set of peers interacting over an asynchronous communication model (Fig. 1). We do not require the model to be existentially bounded, that is, the proposed approach decides realizability even if it is not known a priori whether the specification can be realized with finite buffers, and if it can, for what buffer sizes. We present here a solution that makes this check decidable if the system is well-formed, i.e., (1) in each state a peer can either send one message to a buffer (which we will call a channel), read a message from one channel, or non-deterministically choose between one or more internal actions; and (2) the system is activated by a request from the environment, and a new request is not emitted unless the previous action is completed. Both conditions allow for a class of realistic systems, e.g., peers including a choice among several emissions or receptions, while excluding the class of undecidable systems. We will show how to model such behaviours in Section 3. Condition (1) means that non-deterministic choice is made explicit and excludes race conditions. Condition (2) typically corresponds to service-based systems in which a client (the environment) submits a request and a set of services interact together until returning a response to this request.

Our approach consists of two main steps. First, we explore a sub-behavior — called the canonical schedule — of the possibly infinite state space of the peers interacting via channels. As the canonical schedule may be infinite, only a finite part of it is explored to decide whether in spite of (uncontrollable) internal choices there exists a bounded execution. This check relies on [9], and verifies whether the canonical schedule computed from the set of peers given as input is bounded. If such a bounded execution does not exist, the choreography is not realizable. Otherwise, in a second step, we check realizability by comparing the behaviors of the choreography specification with the previously constructed finite sub-behavior of the peers.

The rest of this paper is organized as follows: Section 2 introduces peers, conversation protocols and our running example. Section 3 presents our approach to checking realizability. Section 4 compares our proposal to related work, and Section 5 ends the paper with some concluding remarks.

2 Peers and Conversation Protocols

In this section, we present the notations we use in the rest of this paper to specify choreographies and peers.
Peers are described using Labelled Transition Systems (LTSs). Peers interact by message passing through point-to-point channels. In this paper, we consider asynchronous communication where each peer is equipped with one channel for each type of message the peer can receive from a given sending peer. In the peer transition systems, write and read actions to and from a channel are written $ch!$ and $ch?$, respectively.

**Definition 1 (Peer).** A peer is a Labelled Transition System (LTS) $P = (S, s^0, \Sigma, T)$ where $S$ is a finite set of states, $s^0 \in S$ is the initial state, $\Sigma = \Sigma^! \cup \Sigma^? \cup \Sigma^\text{int}$ is a finite alphabet partitioned into a set of sending, receiving, and choice actions (internal actions), and $T \subseteq S \times \Sigma \times S$ is a transition relation.

A peer can either send on a channel $ch$ with action $ch! \in \Sigma^!$, read from a channel $ch$ with $ch? \in \Sigma^?$, or choose among one or more internal actions $a \in \Sigma^\text{int}$. Final states are not made explicit and correspond to states without outgoing transitions.

A conversation protocol is an LTS specifying the desired set of conversations from a global point of view. Each transition specifies an interaction between two peers $P_s$, $P_r$ on a specific channel $ch$. A conversation protocol makes explicit the application order of interactions. Sequence, choice, and loop are modeled using a sequence of transitions, several transitions going out from a same state, and a cycle in the LTS, respectively.

**Definition 2 (Conversation protocol).** A conversation protocol $CP$ for a set of peers $P_i$, $i \in \{1, \ldots, n\}$ is an LTS $CP = (S, s^0, L, T)$ where a label $l \in L$ is a tuple $(j, k, ch)$ where $P_j$ and $P_k$ are the sending peer and receiving peer, respectively, $P_j \neq P_k$, and $ch$ is a channel on which those peers interact. We require that each channel has a unique sender and receiver: $\forall (i, j, ch), (i', j', ch') \in L : ch = ch' \implies i = i' \land j = j'$.

**Running example** In this paper, for illustration purposes, we use a bug report repository involving four peers: a client or environment (env), a bug report interface (int), a database (db), and a counter (c). We give successively a conversation protocol (Fig. 2) describing the requirements, that is what the
designer expects from the composition-to-be, and four candidate peers (Fig. 3). The conversation protocol starts with a login interaction between environment and interface, followed by the submission of a bug. Then, interface sends the bug to database to store it, and interacts with counter which stores the number of submitted bugs. Finally, database sends a bug identifier which is forwarded by interface to environment. Interactions in Figure 2 are written using exponent notation, e.g., $\text{submit}^\text{env, int}$ stands for $(\text{env, int, submit})$.

![Fig. 2. Running example: conversation protocol](image)

Figure 3 shows four peers that are candidate to a distributed implementation of our conversation protocol example. For instance, interface receives login information (login?) and a bug (submit?) from environment, sends the bug to database (store!), interacts with counter (count!), receives the identifier from database (ident?), and finally sends the acknowledgement to environment (ack!).

![Fig. 3. Running example: peers (A) environment, (B) database, (C) interface, (D) counter](image)

Although these peers seem to implement the conversation protocol, it is hard by visual analysis only to claim whether this is the case or not, even for such a simple example. Moreover, since we assume an asynchronous communication model, deciding whether the conversation protocol can be implemented by the
peers communicating through bounded buffers, is in general non-trivial. In the rest of this paper we propose an automated technique to check whether a conversation protocol is bounded-realizable by a system of interacting peers.

3 Checking Bounded Realizability

In this section, we present the different steps of our method to check whether a set of peers interacting asynchronously implements a (centralized) conversation protocol. It works in two successive steps. First, we analyse the canonical schedule generated from the peer composition using results presented in [9]. If the schedule is finite, we check realizability by comparing the behaviors of the conversation protocol with the schedule. Otherwise, the conversation protocol is not bounded-realizable by the system of communicating peers.

Definition 3 (Asynchronous product). The asynchronous product of a set of peers $P_i = (S_i, s^0_i, \Sigma_i, T_i)$ is the peer $P_1 || ... || P_n = (S, s^0, \Sigma, T)$ where $S = S_1 \times ... \times S_n$, $s^0 = (s^0_1, ..., s^0_n)$, $\Sigma = \bigcup_i \Sigma_i$, and

$$T = \{ ((s_1, ..., s_n), a, (s'_1, ..., s'_n)) \mid \exists i : (s_i, a, s'_i) \in T_i \land \forall j \neq i : s'_j = s_j \}$$

A composite is a set of peers communicating through emissions and receptions over a set of point-to-point channels.

Definition 4 (Composite). A composite is a tuple $(P, Ch)$ of a set $P = \{ P_i \mid i = 1, ..., n \}$ of peers $P_i = (S_i, s^0_i, \Sigma_i, T_i)$ equipped with a set of channels $Ch = \{ ch_i \}$. We require that $\Sigma^i_1 \cap \Sigma^j_1 \neq \emptyset$ $\Rightarrow$ $i = j$ and $\Sigma^i_1 \cap \Sigma^j_2 \neq \emptyset$ $\Rightarrow$ $i = j$, that is, each channel has a unique reader and writer. Furthermore, we assume that $\Sigma^i_1 \cap \Sigma^i_2 = \emptyset$ for all $i$, that is, each channel links two different peers.

From a conversation protocol $CP$ we can compute a composite where each peer is obtained by making abstraction from all other peers in $CP$, and keeping the same channels as in $CP$.

Definition 5 (Projection). The composite obtained by translation of a conversation protocol $CP = (S, s^0, L, T)$ over channels $Ch$ is a tuple $\pi(\text{CP}) = (\{ P_i \}, Ch)$ where $P_i = (S_i, s^0_i, \Sigma_i, T_i)$ is the LTS obtained by replacing in $CP$ each action label $(p, q, ch) \in L$ with $ch^1$ if $p = i$; with $ch^2$ if $q = i$; and with $\tau$ (internal action) otherwise, and finally removing the $\tau$-transitions by applying the standard determinization algorithms [15].

By Definition 2 it can be shown that $\pi(\text{CP})$ satisfies the requirements of Definition 4 that each channel has a unique reader and writer, and both are different.

Example 1. We show in Figure 4 the peer database obtained by projection from the conversation protocol presented in Section 2. The final peer (right-hand side) is obtained by determinization and minimization of the left-hand side peer.
A configuration of a set of channels $Ch = \{ch_1, ..., ch_n\}$ is a vector in $\mathbb{N}_{\geq 0}^n$ of non-negative integers associating with each channel the number of buffered messages. Let $0$ denote the tuple of $n$ empty channels.

**Definition 6 (Semantics of a composite).** The semantics of a composite $C = (\{P_i\}, Ch)$ with $\|P_i\| = (S, s^0, \Sigma, T)$ is the LTS $\text{sem}(C) = (Q, q^0, \Sigma, \rightarrow)$ where $Q = S \times \mathbb{N}^{\|Ch\|}$, $q^0 = (s^0, 0)$, and $\rightarrow \subseteq Q \times \Sigma \times Q$ is the least transition relation satisfying the following rules:

\[
\begin{align*}
\frac{(s, ch_k!, s') \in T}{(s, (c_1, ..., c_k, ..., c_n)) \xrightarrow{ch_k!} (s', (c_1, ..., c_k + 1, ..., c_n))} \quad (\text{SND})
\end{align*}
\]

\[
\begin{align*}
\frac{(s, ch_k?, s') \in T \quad c_k \geq 1}{(s, (c_1, ..., c_k, ..., c_n)) \xrightarrow{ch_k?} (s', (c_1, ..., c_k - 1, ..., c_n))} \quad (\text{RCV})
\end{align*}
\]

\[
\begin{align*}
\frac{(s, a, s') \in T \quad a \in \Sigma^{\text{nat}}}{(s, c) \xrightarrow{a} (s', c)} \quad (\text{INT})
\end{align*}
\]

For a tuple $b = (b_i)_{ch_i \in Ch}$ of channel bounds, let $\text{sem}(C)/b = (Q', q^0, \Sigma, \rightarrow')$ with

\[
Q' = \{ (s, (c_1, ..., c_n)) \in Q \mid \forall i = 1, ..., n : c_i \leq b_i \}
\]

and $\rightarrow' = \{ (q, a, q') \in \rightarrow \mid q, q' \in Q' \}$ be the sub-graph of $\text{sem}(C)$ restricted to the states satisfying the buffer bounds.

For a state $q \in Q$, let $\text{enabled}(q)$ be the set of actions $a \in \Sigma$ such that $q \xrightarrow{a} q'$ for some $q'$.

We now define when a composite implements a conversation protocol. The composite can be obtained by projection of the conversation protocol, or by assembling existing (off-the-shelf) peers.

**Definition 7 (Implements, $\models_b$).** Given a conversation protocol $CP = (S, s^0, L, T)$ over peers $1, ..., n$ and a set of channels $Ch$, a composite $C = (\{P_i\} \mid i = 1, ..., m), Ch'$ with $m \geq n$ and $P = \|\{P_i\} \mid i = 1, ..., m\|$ over alphabets $\Sigma_i$, and $G = (Q, q^0, \Sigma, \rightarrow)$ a sub-graph of $\text{sem}(C)$, let $\preceq \subseteq Q \times S$ be the greatest relation $\prec$ such that if $q \prec s$ then:

\[
\text{Fig. 4. Peer database generated by projection: (left) before and (right) after determinization and minimization}
\]
1. If \((s, (i, j, ch), s') \in T\) then \(\exists k \geq 0 \exists q_1, \ldots, q_k \in Q \exists a_1, \ldots, a_k \in \Sigma \setminus C\left\{ \right\} \) 
with \(q \xrightarrow{a_1} q_1 \xrightarrow{a_2} \ldots \xrightarrow{a_k} q_k \xrightarrow{ch!} q'\) with \(ch! \in C\) if there exists a tuple of bounds \(b = (b_i)_{ch \in Ch}\) on the channels such that \(C \models b CP\).

2. If \(q \xrightarrow{a} q'\) with \(a \in \Sigma \setminus C\left\{ \right\}\) then \(q' \prec s\) (communication in \(CP\))

3. If \(q \xrightarrow{a} q'\) with \(a \in \Sigma \setminus C\left\{ \right\}\) then \(q' \prec s\) (unobservable transition of \(C\))

where \(\Sigma^{\text{C}\setminus \text{CP}} = \{ch! \in \Sigma \mid ch \notin Ch\} \cup \Sigma^\tau \cup \Sigma^\text{int}\).

Given a tuple \(b = (b_i)_{ch \in Ch}\) of channel bounds, \(C\) implements \(CP\) under \(b\), written \(C \models_b CP\), if \(\text{sem}(C)/b \preceq CP\).

Intuitively, the conversation protocol and the composite must be bisimilar with respect to the communication over channels in \(Ch\). The composite may encompass additional peers and use auxiliary channels that are not part of the conversation protocol, and execute internal actions. Other notions of implementation could have been chosen such as weaker notions [17] or notions taken receptions into account as well [21].

Remark 1. \(\pi(CP) \not\models CP\), in general, as \(\pi(CP)\) may have more behaviors than \(CP\). Some solutions exist that either propose well-formedness rules to enforce the choreography specification to be realizable [7], or extend the choreography language with new constructs (named dominated choice and loop) that make the peers obtained by projection respect the choreography specification [22]. However, these approaches focus on synchronous communication and do not provide any solution to the boundedness issue inherent to asynchronous communication.

Example 2. If we compare, using Definition 7, the execution traces that can be produced from the conversation protocol given in Figure 2 with those executed by the composite consisting of the peers presented in Figure 3, this check says that the composite does not implement the conversation protocol because the trace login!, login?, submit!, submit?, store!, store?, ident! belongs to the composite but is not a valid trace for the conversation protocol. Indeed, the latter specifies that the interaction between interface and counter (\(\text{count}_{\text{int}, c}\) in Figure 2) must occur before database sends its response to interface. However, this cannot be imposed according to the different peers we reuse for implementation purposes. To work this out, the designer has two possible choices: (i) to relax the choreography specification constraints by making explicit that \(\text{count}_{\text{int}, c}\) and \(\text{ident}_{\text{db}, \text{int}}\) can be executed in any order (this would be specified using a diamond of interleaved transitions in the conversation protocol), or (ii) to use extra synchronizations such as those proposed in [23] to enforce peers to respect the ordering constraints specified in the conversation protocol.

Definition 8 (Bounded-realizable). A conversation protocol \(CP\) is bounded-realizable by a composite \(C = (\{P_i\}, Ch)\) if there exists a tuple of bounds \(b = (b_i)_{ch \in Ch}\) on the channels such that \(C \models_b CP\).
For a given composite, the existence of a non-blocking quasi-static scheduler that ensures boundedness of the channels in spite of uncontrollable non-determinism of the peers, has shown to be undecidable in general [9]. The goal of the remainder of this section is to define a decidable subclass of composites and effectively decide, for a system of this class, whether a conversation protocol is bounded-realizable by a set of peers. In order to tackle this question we need some more definitions.

**Definition 9 (Data-branching [9]).** A peer $P = (S, s^0, \Sigma, T)$ is data-branching if for any $s \in S$, one of the following is true:

- All outgoing transitions are choice transitions, and there is at least one such transition (and $s$ is called choice state).
- $s$ has exactly one outgoing transition $(s, a, s')$ and $a \in \Sigma^d$ (and $s$ is a sending state).
- $s$ has at most one outgoing transition $(s, a, s')$ and $a \in \Sigma^r$ (and $s$ is a polling state).

In particular, a state without any outgoing transition is a polling state.

Intuitively, the data-branching assumption ensures that non-determinism in the global behavior only comes from internal choice and not from race conditions caused by simultaneous listening on several channels, or non-deterministic emission to several channels. The transitions issued from choice states can be seen as the non-deterministic choice obtained from conditional branching after making abstraction from data. Ruling out concurrently enabled emissions is not a restriction, due to the asynchronous model of communication. Figure 5 shows how a choice state can be used to encode non-deterministic emissions: each emission is preceded by a choice transition (this pattern corresponds to an internal choice in process algebra, see CSP [14] for instance).

![Fig. 5. Modeling non-deterministic emissions](image)

Next we define round-separation of a composite, ensuring that a new request triggering a reaction of the composite is not emitted unless the previous reaction of the composite is completed.

**Definition 10 (Round-separated).** A composite $C = (\{ P_i \}, Ch)$ of peers $P_i = (S_i, s^0_i, \Sigma_i, T_i)$ with $\text{sem}(C) = (Q, q^0, \Sigma, \rightarrow)$ is round-separated if
1. there exists some peer $P_k$ and action $\text{init} \in \Sigma_k$ such that $\text{enabled}(q^0) = \{\text{init}\}$;
2. $\forall q = ((s_1, \ldots, s_n), c) \in Q : \text{init} \in \text{enabled}(q) \implies \forall j \neq k : s_j$ is a polling state and $c = 0$; and
3. from any reachable state of $C$, $\text{final}(C) = \{q = (s, c) | \text{enabled}(q) \subseteq \{\text{init}\} \land \ c = 0\}$ is reachable.

In a round-separated composite the only action enabled in $q^0$ — call it $\text{init}$ — is enabled only in states $q = (s, c) \in Q$ where all other peers are polling in states and all channels are empty. The set $\text{final}(C)$ is the set of final states where at most $\text{init}$ is enabled, all other peers are in polling states, and all channels are empty.

Example 3. The composite consisting of the peers presented in Figure 3 is round-separated: there is some $\text{init}$ action (login!) initiating the interaction process (condition 1 in Definition 10), this action is never reached again (therefore condition 2 does not need to be verified), and from any reachable state in the composite state, a final state with empty channels is reachable where all peers are in polling states (condition 3 in Definition 10), see the shaded states in Figure 6.

As the requirement of Definition 10 is expressed on the semantics of $C$, we have two ways to effectively check it: by using some syntactic check that is a sufficient but not necessary condition, or on-the-fly during the state-space exploration. In the approach presented here we choose the second option.

Definition 11 (Well-formed). A composite is well-formed if it is round-separated and its peers are data-branching.

The condition of well-formedness allows us to leverage the results of [9] to effectively decide whether a conversation protocol is bounded-realizable. For the
sake of a self-contained presentation we cite the following definitions, slightly adapted from [9] to match our framework.

Definition 12 ($P_{\text{poll}}^{i}$, $P_{\text{choice}}^{i}$, $P_{\text{send-min}}^{i}$). Given a composite $C = \{P_{1}\}$, $Ch$ of data-branching peers and a state $q = (s, c)$ with $s = (s_{1}, ..., s_{n})$ and $c = (c_{1}, ..., c_{n})$ of $\text{sem}(C)$, let $P_{\text{poll}}^{i}$, $P_{\text{choice}}^{i}$, and $P_{\text{send}}^{i}$ be the sets of indices of the peers that are in a polling state, a choice state, and a sending state, respectively. Let $P_{\text{send-min}}^{i} \subseteq P_{\text{send}}^{i}$ be the set of indices $i$ such that $ch_{k} \in \Sigma_{i}∩\text{enabled}(q)$ with $c_{k} = \min\{c_{j}\}$ be the subset of peers ready to send a message to a channel holding a minimal number of messages.

The basic idea of a canonical schedule is to constrain the execution of a composite by giving priority to read and choice actions over write actions. In the case where only write actions are enabled, one of those writing to a channel containing a minimal number of messages is chosen.

Definition 13 (Canonical schedule). Given a composite $C = \{P_{1}\}$, $Ch$ with $P_{1} = (S_{1}, s_{1}^{0}, \Sigma_{1}, T_{1})$ and $\|P_{1} = (S, s^{0}, \Sigma, T)$, the canonical schedule of $C$ is the least sub-graph $CS(C) = (Q_{\text{ca}}, q^{0}, \Sigma, \rightarrow_{\text{ca}})$ of $\text{sem}(C) = (Q, q^{0}, \Sigma, \rightarrow)$ such that $q^{0} \in Q_{\text{ca}}$ and for any $q = (s, c) \in Q_{\text{ca}}$ with $s = (s_{1}, ..., s_{n})$:

- If $P_{\text{poll}}^{i} \cup P_{\text{choice}}^{i} \neq \emptyset$ and $q \xrightarrow{a} q'$ with $q' = (s', c')$, $s' = (s'_{1}, ..., s'_{n})$, and $a \in \Sigma_{1}^{\text{int}} \cup \Sigma_{k}^{\text{int}}$ where $k = \min P_{\text{poll}}^{i} \cup P_{\text{choice}}^{i}$, then $q \xrightarrow{a} q'$.
- Otherwise, if $q \xrightarrow{a} q'$ with $a \in \Sigma_{1}^{\text{int}} \cup \Sigma_{k}^{\text{int}}$ where $k = \min P_{\text{send-min}}^{i}$, $q' = (s', c')$, $s' = (s'_{1}, ..., s'_{n})$, and $(s_{k}, a, s_{k}) \in T_{k}$, then $q \xrightarrow{a} q'$.

As the canonical schedule may be infinite, an order between prefixes is defined next that will be used to explore only a finite part of the potentially infinite state space of a composite.

Given an LTS $(S, s^{0}, \Sigma, \rightarrow)$, states $q, q' \in S$, and a sequence $\sigma = a_{1}a_{2}...a_{n} \in \Sigma^{*}$, we write $q \xrightarrow{\sigma} q_{n}$ if there are states $q_{1}, ..., q_{n-1} \in S$ such that $q \xrightarrow{a_{1}} q_{1} \xrightarrow{a_{2}} ... \xrightarrow{a_{n}} q_{n}$.

Definition 14 ($\prec_{ca}$). Let $\sigma, \sigma' \in \Sigma^{*}$ with $q^{0} \xrightarrow{\sigma} (s, c)$ and $q^{0} \xrightarrow{\sigma'} (s', c')$. Define $\sigma \prec_{ca} \sigma'$ if all of the following conditions hold:

1. $\sigma$ is a prefix of $\sigma'$
2. $s = s'$ and $\forall ch \in Ch$, $c(ch) \leq c'(ch)$
3. there exists some $ch \in Ch$ such that
   - $\sigma = \sigma_{1}ch$ for some $\sigma_{1} \in \Sigma^{*}$ with $\text{max}(\sigma_{1}) < \text{max}(\sigma)$; and
   - $\sigma' = \sigma_{2}ch$ for some $\sigma_{2} \in \Sigma^{*}$ with $\text{max}(\sigma_{2}) < \text{max}(\sigma')$

where

$$\text{max}(\sigma) = \max\{\max\{c_{1}, ..., c_{n} | q^{0} \xrightarrow{\sigma'} (s, (c_{1}, ..., c_{n}))\} | \sigma' \text{ is a prefix of } \sigma\}$$
Algorithm 1 (Decision procedure) Given a composite \( C = (\{P_i\}, Ch) \) with \( \|P_i = (S, s^0, \Sigma, T) \) and \( \text{sem}(C) = (Q, q^0, \Sigma, \rightarrow) \), we construct a finite coverability tree \([9]\) \( \text{Tr}(C) \subseteq \Sigma^* \) as follows. First, \( \varepsilon \in \text{Tr}(C) \). For any \( \sigma \in \text{Tr}(C) \) and \( a \in \Sigma \) with \( q^0 \xrightarrow{\sigma} q_\sigma \) and \( q_\sigma \xrightarrow{a} q' = (s, c) \):

- If \( \text{init} \in \text{enabled}(q') \) and either \( |\text{enabled}(q')| \geq 2 \) or \( c \neq 0 \) then \( C \) is not round-separated; stop.
- Otherwise, if there exists \( \sigma' \in \text{Tr}(C) \) such that \( \sigma' \prec_{ca} \sigma \alpha \) then \( C \) is unbounded or \( \text{final}(C) \) is unreachable; stop.
- Otherwise, if there is no \( \sigma' \in \text{Tr}(C) \) such that \( q^0 \xrightarrow{\sigma'} q_\sigma \) then add \( \sigma \alpha \) to \( \text{Tr}(C) \).

It can be shown that Algorithm 1 terminates, since either the canonical schedule is finite and all states have been explored, or there are two prefixes \( \sigma, \sigma' \) such that \( \sigma \prec_{ca} \sigma' \) \([9]\).

Example 4. We give in Figure 7, the canonical schedule generated from the composite given in Figure 3 by application of Definition 13 (peers are ordered \( \text{wrt.} \) their alphabetical identifiers A, B, C, D). One can see that the choice made by the environment is present in the canonical schedule and three possible behaviours are derived.

![Diagram](image-url)

**Fig. 7.** Running example: canonical schedule (top) where \( tr \) is the transition sequence shown on bottom

The canonical schedule is unbounded: if the environment decides to submit several bugs without consuming acknowledgements (submit branch in the peer environment), then by applying Algorithm 1, we can generate traces from the
canonical schedule where the channel size increases (in particular, the size of the channel in the peer environment storing acknowledgements), and case 2 of this algorithm detects this unboundedness case. A solution to this issue is to use a peer environment’ which systematically consumes acknowledgements sent by the peer interface, as in Figure 8. If we use this new peer environment’ and the other peers presented in Figure 3, the corresponding canonical schedule (given in Figure 9) is bounded because each channel is read immediately after being written.

![Fig. 8. A candidate peer environment’ avoiding the unboundedness issue](image1)

![Fig. 9. Example: canonical schedule obtained with the peer environment’ given in Figure 8](image2)

**Theorem 1 (Bounded schedule).** Consider a conversation protocol $CP$ and a composite $C$ composed of data-branching peers.

1. If $C$ is round-separated then Algorithm 1 does not terminate with a negative round-separation result.
2. Otherwise, if Algorithm 1 terminates with a negative boundedness or reachability result, then $CP$ is not bounded-realizable by $C$.
3. If Algorithm 1 terminates without a negative result (round-separation, boundedness, or reachability), then $C$ is well-formed and $CS(C)$ is finite.

**Proof.** 1. The claim follows directly from Definition 10.
2. If Algorithm 1 terminates with a negative boundedness or reachability result, then the non-boundedness of $C$ or unreachability of $final(C)$ follows from Proposition 8 of [9]. The only difference in our setting is that we explicitly model resets in the form of $init$ transitions. As by hypothesis of this item, $C$ is round-separated, $final(C)$ is reachable by Definition 10. Therefore, $C$ is unbounded, and the claim follows.

3. Round-separation under the canonical schedule is ensured by the normal termination of the algorithm. Round-separation on arbitrary runs is obtained by a reordering argument similar to that used in [9]. Well-formedness then follows directly from the hypothesis of data-branching peers and round-separation.

Example 5. If we consider the peer environment given in Figure 8 and the three other peers presented in Figure 3, Algorithm 1 terminates with a positive result, meaning that the composite is well-formed and the canonical schedule is finite (see Figure 9).

Notice that even if the canonical schedule $CS(C)$ of a composite $C$ is finite, the semantic graph $semi(C)$ may still be infinite. However, if $C$ is well-formed and $CS(C)$ is finite, then bounded-realizability of a conversation protocol $CP$ by $C$ can be effectively verified.

Theorem 2 (Bounded-realizability). Given a conversation protocol $CP$ and a well-formed composite $C$, $CP$ is bounded-realizable by $C$ if and only if the canonical schedule $CS(C) = (Q, q^0, \Sigma, \rightarrow)$ is finite and $C \models_b CP$, where $b = (b_i)_{ch \in Ch}$ with $\forall i, b_i = \max \{ c_i | \exists s, c = (c_1, ..., c_n) : q^0 \rightarrow^* (s, c) \}$, and $\rightarrow^*$ denotes the reflexive and transitive closure of $\rightarrow$.

That is, $CP$ is bounded-realizable by $C$ if and only if it is bounded-realizable for channel bounds used by the canonical schedule.

Proof. (sketch) “If”: if $C \models_b CP$ then clearly, $CP$ is bounded-realizable by $C$.

“Only if”: suppose that $CP$ is bounded-realizable by $C$, say $C \models_{b'} CP$ for some tuple $b'$ of buffer bounds. Then $b' \geq b$ by construction of the canonical schedule. In particular, $CS(C)$ is finite. Moreover, $semi(C)/b'$ is a sub-graph of $semi(C)/b$. Therefore, it can be shown by structural induction that items 2. and 3. of Definition 7 still hold for $semi(C)/b$. Moreover, as $C$ is well-formed and thus round-separated, $final(C)$ is reachable from any reachable state of $semi(C)$. This ensures that all pending write actions will eventually be executed, such that item 1. of Definition 7 is still satisfied. It follows that $C \models_b CP$.

Example 6. Although the canonical schedule generated from the peer environment given in Figure 8 and the peers database, interface, and counter presented in Figure 3 is finite, the corresponding semantic graph is infinite because the counter has no obligation to read. To check bounded-realizability, the required channel size is one for all channels since each channel can be read immediately after being written. If we consider an extension of the conversation protocol given in Figure 2 where count$^{int,c}$ and return$^{db,int}$ can be interleaved — as discussed...
in Example 2 —, then this conversation protocol is bounded-realizable by the composite.

4 Related Work

The realizability results we present in this paper rely on [9] where the authors identify a decidable class of systems consisting of non-deterministic communicating processes that can be scheduled while ensuring boundedness of buffers. There has been quite some work on the analysis of infinite communication buffers in concurrent systems. Abdulla et al. [1] proposed some verification methods for Communicating Finite State Machines. They showed the decidability and provided algorithms for verification (safety and some forms of liveness properties) of lossy channel systems. A sufficient condition for the unboundedness of communication channel was proposed in [16]. In [18, 19], the authors present an incomplete boundedness test for communication channels in Promela and UML RT models. They also provide a method to derive upper bound estimates for the maximal occupancy of each individual message buffer. More recently, [10] proposed a causal chain analysis to determine upper bounds on buffer sizes for multi-party sessions with asynchronous communication. Our goal here is to compute the minimal buffer sizes which make the interacting peers realize the choreography, but this does not mean that a bound exists for each buffer. Therefore, the results presented in [18, 19, 10] would not help to solve the problem we tackle here.

Most of the work dedicated to the realizability issue assumes a synchronous communication model, see for instance [6, 20, 7, 22]. In [6, 20], the authors define models for choreography and orchestration, and formalise a conformance relation between both models. The results presented in [7, 22] formalise some well-formedness rules to enforce the specification to be realizable. More precisely, in [7], the authors identify three principles for global description under which they define a sound and complete end-point projection, that is the generation of distributed processes from the choreography description. In [22], the authors propose a choreography language with new constructs (named dominated choice and loop) in order to implement unrealizable choreographies. During the projection of these new operators, some communications are added in order to make peers respect the choreography specification. However, these solutions prevent the designer from specifying what (s)he wants to, and it also complicates the design by obliging the designer to make explicit extra-constraints in the choreography specification, e.g., by associating dominant roles to certain peers.

Only a few works focused on the realizability problem assuming an asynchronous communication model, that is communication using message buffers. Fu et al. [11] proposed three sufficient conditions (lossless join, synchronous compatible, autonomous) that guarantee a realizable conversation protocol. More recently, Sasu and Bultan proposed to check conformance using synchronizability [4]: A set of peers is synchronizable if systems produced on one hand with synchronous communication, and on the other with 1-bounded asynchronous
communication, are equivalent. If a set of peers is synchronizable, one can check whether it is conformant to a choreography using existing finite state verification tools. However, if one of the conditions in [11] or synchronizability is not satisfied, nothing can be concluded. Our approach works for systems that are not synchronizable.

Bultan and Fu [5] defined some sufficient conditions to test realizability of choreographies specified with collaboration diagrams (CDs). In [23], the authors refine and extend this former work with some techniques to enforce realizability (by adding additional synchronization messages among peers), and a tool-supported approach to automatically check the realizability of CDs for bounded asynchronous communication. The realizability problem for Message Sequence Charts (MSCs) has also been studied (see for instance [2, 26, 3]). For example, [3] presents some decidability results on bounded MSC graphs, that are basically graphs obtained from MSCs using bounded buffers. These solutions are limited because branching and cyclic behaviours are not well supported by CDs and MSCs (no choice in CDs, no cyclic behaviours in MSCs, and only loops on a same message in CDs).

Lohmann and Wolf [21] show how realizability of choreography automaton can be verified by using existing techniques for the controllability problem, which checks whether a service has compatible partner processes. Their approach works for peers interacting via arbitrary bounded buffers, and only consider finite conversations, whereas we can handle infinite state space systems.

Genest et al. [13] establish equivalence of existentially bounded communicating automata with globally cooperative compositional message sequence graphs and monadic second-order logic.

In [8] on quasi-static scheduling of free-choice Petri nets, a coverability criterion is defined whose function, similar to the relation $≺_{ca}$, is to explore only a finite part of a potentially infinite state space. The authors conjecture completeness of the criterion. Based on [8], [25] uses discrete controller synthesis to automatically construct converters between peers so as to ensure bounded buffering and deadlock freedom.

Compared to all these works, our approach provides a check for realizability under asynchronous communication, and goes beyond most results which assume arbitrary bounded buffers, this check being undecidable for unbounded buffers. Here, we rely on a boundedness analysis of the peer composition, and provide a decidable technique for well-formed systems of communicating peers. We also extend existing results for conversation protocol realizability by considering peer composition (e.g., those which are not synchronizable) for which existing solutions [11, 4] cannot conclude anything.

5 Concluding Remarks

In this paper, we have presented an approach for checking whether a conversation protocol can be implemented by a set of distributed peers interacting asynchronously. The realizability check relies on the boundedness of the canon-
ical schedule computed from the candidate peers. If this schedule is infinite, the conversation protocol cannot be realized with bounded buffers by the peers. If this schedule is finite, we compare the LTS obtained from the conversation protocol with the LTS generated from the peer composition to check whether these peers implement the choreography specification.

An interesting direction of future work we intend to study is the generalization of our framework to multi-session protocols. This will require several generalizations to our results, in particular extending the modeling formalism and refinement relation, and relaxing the round-separation requirement.

References


Abstract. We present an automata theoretic framework for modular schedulability analysis of networks of real-time asynchronous actors. In this paper, we use the coordination language Reo to structure the network of actors and as such provide an exogenous form of scheduling between actors to complement their internal scheduling. We explain how to avoid extra communication buffers during analysis in some common Reo connectors. We then consider communication delays between actors and analyze its effect on schedulability of the system. Furthermore, in order to have a uniform analysis platform, we show how to use UPPAAL to combine Constraint Automata, the semantic model of Reo, with Timed Automata models of the actors. We can derive end-to-end deadlines, i.e., the deadline on a message from when it is sent until a reply is received.

1 Introduction

Schedulability analysis in a real-time system amounts to checking whether all tasks can be accomplished within the required deadlines. In a client-server perspective on distributed systems, tasks are created on a client, sent to the server (e.g., as a message), and then finally performed on the server. A deadline given by the client for a task covers three parts: the network delay until the message reaches the server, the queuing time until the task starts executing, and the execution time. In case a reply is sent back to the client, an end-to-end deadline also includes the network delay until the reply reaches the client and is processed.

In previous work [14–16], we employed automata theory to provide a modular approach to the schedulability analysis of real-time actor models, assuming direct and immediate communication between actors, i.e., zero communication delays. An actor [1,12] (à la Rebeca [25]) is an autonomous entity with a single thread of execution. Actors communicate by asynchronous message passing, i.e., incoming messages are buffered and the code for handling each message is defined in a corresponding method. We model each method as a timed automaton [3] where a method can send messages while computation is abstracted in passage of time. In our framework, an actor can define a local scheduler and thus reduce the

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Section 2 explains a modular way to analyze a system of actors. To be able to do so, the expected usage of each actor is specified in a separate timed automaton, called its behavioral interface; this is a contract between the actor and its environment [22], which among other things, includes the schedulability requirements for the actor in terms of deadlines. Every actor is checked individually for schedulability with regard to its behavioral interface. We showed in [15] that schedulable actors need finite buffers; the upper-bound on buffer size can be computed statically. When composing a number of individually schedulable actors, the global schedulability of the system can be concluded from the compatibility of the actors [16]. Being subject to state-space explosion, we gave a technique in [16] to test compatibility.

The contribution of this paper is twofold. First in Section 3, we extend the above framework with Reo [4] to enable exogenous coordination of the actors. This provides a separation of concerns between computation and coordination. Reo can be used as a “glue code” language for compositionally building connectors that orchestrate the cooperation between components or services in a component-based system or a service-oriented application. An important feature of Reo is that it allows for anonymous communication, i.e., the sender of a message does not need to know the recipient; instead the Reo connector will forward the message to the proper receiver.

With Reo, individually schedulable actors can be used as off-the-shelf modules in a wider variety of network structures. This requires a new compatibility check for our analysis that incorporates the Reo connectors. Our extension preserves the asynchronous nature of the actors, therefore the Reo connectors must have a buffer at every input/output node, which may lead to state-space explosion. To avoid this problem, we provide techniques to optimize the analysis by reusing internal actor buffers in the Reo connectors that are single-input and/or single-output. We show that in this approach the upper-bound on the size of the buffers of the schedulable actors need not be increased. In Section 5, we give examples of other Reo connectors that can take advantage of the same optimization technique. In any case, we assume coordination and data flow by Reo happens in zero time.

As our second contribution, we analyze in Section 4 the effect of communication delays on the schedulability of a distributed system. For simplicity in presentation, we assume no coordination with Reo in this section. The communication medium between every pair of actors is modeled abstractly by a fixed delay value, called their distance. We first describe how to implement the effect of delay on messages in an efficient manner with respect to schedulability analysis. Secondly we extend the compatibility check to take message delays into account. The latter is non-trivial because sending and receiving messages do not happen at the same time any more. Nevertheless, this complication can be hidden from the end user by implementing it in an automatic test-case generation algorithm.
We argue in Section 5 that coordination with Reo and communication delays are orthogonal and can be combined.

As a running example, we consider a client/server composition of two actors. Assuming that the client is faster, the overall system would not be schedulable because the server would not be able to respond in time. This situation can be remedied by using Reo to connect the client to multiple server instances in order to compensate for their slowness. Nonetheless, the client still thinks it is communicating with one server, i.e., coordination is transparent to the client and the server actors. In other words, modularity of the analysis is preserved.

1.1 Related Work

Schedulability has usually been analyzed for a whole system running on a single processor, whether at modeling [2, 10] or programming level [7, 18]. We address distributed systems modeled as a network of actors (connected by Reo circuits) where each actor has a dedicated processor and scheduling policy.

The work in [11] is also applicable to distributed systems but is limited to rate monotonic analysis. Our analysis being based on automata can handle non-uniformly recurring tasks as in Task Automata [10]. In Task automata, however, a task is purely specified as computation times and cannot create sub-tasks.

In our approach, behavioral interfaces are key to modularity. A behavioral interface models the most general message arrival pattern for an actor. The behavioral interface can be viewed as a contract, as in ‘design by contract’ [22], or as a most general assumption in modular model checking [20] (based on assume-guarantee reasoning). Schedulability is guaranteed if the real use of the actor satisfies this assumption.

RT-Synchronizers [23] also provide some sort of coordination among actors, however, they are designed for declarative specification of timing constraints over groups of untimed actors. Therefore, they do not speak of schedulability of the actors themselves; in fact, a deadline associated to a message is for the time before it is executed and therefore cannot deal with the execution time of the task itself or sub-task generation.

In [9, 14], our approach is extended to accommodate synchronization statements and replies of the Creol language [17]. Asynchronous message passing in Creol is augmented with explicit return values and message synchronization. Therefore, Creol has the natural means to model end-to-end deadlines, however the work in [9, 14] does not support network delays. In present work, an end-to-end deadline including network delays can be computed manually by adding up the deadlines of the message and its corresponding reply message.

There are several coordination languages that can be used to coordinate actors, two of which are worth mentioning. First there is the ARC model [24], which aims at coordinating resource usage and QoS goals, and is based on state transition systems. Secondly there is the PBRD model [21], which aims at logical communication behavior, and is based on rewriting logic. Apart from modeling capabilities, unlike the two above, Reo has automata based semantics which allows us to connect naturally to our automata-theoretic framework in [15].
Several semantic models have been suggested for Reo in order to handle data-transfer delays, e.g. [19]. None of these models are yet able to consider the delay in setting up a connection in a distributed way. Therefore in this work, we restrict to centralized Reo connectors and we assume that coordination happens in negligible time. This assumption is reasonable when Reo connectors are deployed local to actors. In this paper, we provide no real-time extensions of Reo; although we propose an algorithm to translate some Reo connectors into Timed Automata.

2 Preliminaries: Real-Time Actors

We use automata theory for modular schedulability analysis of actor-based systems [15,16]. An actor consists of a set of methods which are specified in Timed Automata (TA) [3]. This enables us to use existing tools, for example UPPAAL [6], to perform analysis. Each actor should provide a behavioral interface that specifies at a high level, and in the most general terms, how this actor may be used. As explained later in this section, behavioral interfaces are key to modular analysis of actors. Actors specify local scheduling strategies, e.g., based on fixed priorities, earliest deadline first, or a combination of such policies. Real-time actors may need certain customized scheduling strategies in order to meet their QoS requirements. We describe in this section how to model and analyze actors.

Modeling behavioral interfaces A behavioral interface consists of the messages an actor may receive and send; thus it provides an abstract overview of the actor behavior in a single automaton. A behavioral interface abstracts from specific method implementations, the message buffer in the actor and the scheduling strategy.

To formally define a behavioral interface, we assume a finite global set \( \mathcal{M} \) for method names. A behavioral interface \( B \) providing a set of method names \( M_B \subseteq \mathcal{M} \) is a deterministic timed automaton over alphabet \( \text{Act}^B \) such that \( \text{Act}^B \) is partitioned into two sets of actions:

- outputs: \( \text{Act}_O^B = \{ m? | m \in \mathcal{M} \land m \notin M_B \} \)
- inputs: \( \text{Act}_I^B = \{ m(d)! | m \in M_B \land d \in \mathbb{N} \} \)

Notice the unusual use of ! and ? signs; this is to simplify the analysis as will be explained later. The integer \( d \) associated to input actions represents a deadline. A correct implementation of the actor should be able to finish method \( m \) before \( d \) time units.

Example. Fig. 1 depicts the UPPAAL models for behavioral interfaces of two actors that can communicate in a client-server fashion by sending request and reply
Modeling classes
One can define a class as a set of methods implementing a specific behavioral interface. A class \( R \) implementing the behavioral interface \( B \) is a set \( \{(m_1, A_1), \ldots, (m_n, A_n)\} \) of methods, where

- \( M_R = \{m_1, \ldots, m_n\} \subseteq M \) is a set of method names such that \( M_B \subseteq M_R \);
- for all \( i, 1 \leq i \leq n \), \( A_i \) is a timed automaton representing method \( m_i \) with the alphabet \( \text{Act}_i = \{m! | m \in M_R\} \cup \{m(d)! | m \in M \land d \in \mathbb{N}\} \);

Method automata only send messages while computations are abstracted into time delays by using a clock \( c \). Receiving and buffering messages is handled by the scheduler automata (explained below). Sending a message \( m \in M_R \) is called a self call. A self call with no explicit deadline inherits the (remaining) deadline of the task that triggers it (called delegation); in this case the delegate channel must be used.

Classes have an initial method which is implicitly called upon initialization and is used for the system startup. Execution of a method begins after receiving a signal on the start channel and terminates by sending a signal on the finish channel; this way the scheduler can control execution of the methods. Fig. 2 shows an implementation of the methods of our example.

Modeling schedulers
The scheduler for each actor, containing also its message buffer, is modeled separately as a timed automaton (see Fig. 3). The buffer is modeled using arrays in UPPAAL and thus it can be modeled compactly, i.e., without different locations for different buffer states. The scheduler automaton begins with putting an initial message in the buffer via the initialize function.

The scheduler is input-enabled, i.e., it allows receiving any message from any sender on the invoke channel. The buffer stores along each message its sender and deadline. A free clock is assigned to each message and reset to zero upon

Fig. 2. Method implementations for client and server actors.
2.1 Modular Schedulability Analysis

An actor is an instance of a class together with a scheduler. A closed system of actors is schedulable if and only if all tasks finish within their deadlines. We have shown in [15] that schedulable actors do not put more than $\lceil d_{\text{max}}/b_{\text{min}} \rceil$ messages in the buffer, where $d_{\text{max}}$ is the longest deadline for the messages and $b_{\text{min}}$ is the shortest termination time of its method automata. One can calculate the best case runtime for timed automata as shown by Courcoubetis and Yannakakis [8]. Formally, schedulability is defined as follows.

**Definition 1 (System Schedulability).** A closed system of actors is schedulable if and only if none of the scheduler automata can reach the Error location or exceeds the buffer limit of $\lceil d_{\text{max}}/b_{\text{min}} \rceil$.

Thus, schedulability analysis can be reduced to reachability analysis in a tool like UPPAAL. The intrinsic asynchrony of actors and their message buffers practically lead to state-space explosion. Our approach to modular analysis of the actors (as in [15]) combines model checking and testing techniques to overcome this problem. This is done in the two steps described below.
**Individual actor analysis** The methods of an actor can in theory be called in infinitely many ways, which makes their analysis impossible. However, it is reasonable to restrict only to the incoming method calls specified in its behavioral interface. Input actions in the behavioral interface correspond to incoming messages. Incoming messages are buffered in the actor; this can be interpreted as creating a new task for handling that message. The behavioral interface doesn’t capture internal tasks triggered by self calls. Therefore, one needs to consider both the internal tasks and the tasks triggered by the behavioral interface, which abstractly models the acceptable environments. We can analyze all possible behaviors of an actor in UPPAAL by model checking the network of timed automata consisting of its method automata, behavioral interface automaton $B$ and a scheduler automaton. Inputs of $B$ written $m!$ match inputs in the scheduler written $m?$, and outputs of $B$ written $m?$ match outputs of method automata written $m!$. An actor is schedulable w.r.t. its behavioral interface iff the scheduler cannot reach the Error location and does not exceed its buffer limit.

**Compatibility check** Once an actor is verified to be schedulable with respect to its behavioral interface, it can be used as an off-the-shelf component. In this section, we assume that actors communicate directly with no communication delays. As in modular verification [20], which is based on assume-guarantee reasoning, individually schedulable actors can be used in systems compatible with their behavioral interfaces. Schedulability of such systems is then guaranteed. Intuitively, the product of the behavioral interfaces, called $B$, shows the acceptable sequences of messages that may be communicated between actors.

**Definition 2 (Compatibility).** *Compatibility is defined as the inclusion of the visible traces of the system in the traces of $B$ [16], where visible actions correspond to messages communicated between actors.*

Checking compatibility is prone to state-space explosion due to the size of the system; we avoid this by means of testing techniques. A naïve approach could take a trace from the system $S$ as a test case and check whether it exists also in $B$. This test case generation method is not efficient due to the great deal of nondeterminism in $S$. As proposed in [16], we generate test-cases from $B$. A test-case, first of all, *drives* the system along a trace taken from $B$ and thus restricts system behavior. Secondly, it *monitors* the system along this trace checking for any action that is forbidden in $B$ (as a possible witness for incompatibility). To do the monitoring, every communication between different actors has to be intervened by the test-case automaton. Receiving and forwarding these messages in the test-case are separated by a ‘committed’ location so that UPPAAL executes them with no interruption.

**Example.** Fig. 4 shows a test-case that proves the Server and Client implementations in Fig. 2 to be incompatible. This test-case considers one round of expected request-reply scenario. This scenario is captured in the main line of the test-case (leading to PASS verdict). For the sake of simplicity, we only monitor for one forbidden behavior in this test-case which leads to the FAIL verdict: a lack of a
timely reply is captured as sending two requests without an intermediate reply. When executing this test-case, the FAIL location is indeed reachable because the client in Fig. 2 (i.e., its ‘next’ method) is faster than the server (i.e., its ‘request’ method). We show in Section 3 how Reo can bring flexibility in composing actors such that we can remedy this problem; specifically by allowing us to use two servers with one client.

3 Using Reo for Coordination

Reo can help us coordinate the actors to avoid unexpected message-passing scenarios. That is, we can impose a strict communication pattern on the components, e.g., replicating requests and merging replies or ordering the messages. This can be seen as an exogenous scheduler that might be crucial in schedulability of a composed system. An advantage of Reo for us is its automata-theoretic semantic model, namely Constraint Automata (CA). The idea is that CA models of Reo networks have a high potential to be used in combination with Timed Automata models of actors and thus allow us to analyze our models in UPPAAL.

Complex Reo connectors can be composed out of a basic set of channels. Each channel has exactly two ends that have their own unique identities. A channel end can be a source or a sink. Data enters at the source end and leaves the channel through the sink. To build complex connectors, channels are connected by means of nodes (also called ports). A node is like a pumping station that takes the data on one of the incoming ‘sink’ ends and replicates the data onto all of its outgoing ‘source’ ends. Therefore, channels can be connected by: sequential composition where the data flows from one channel to the next one; a non-deterministic choice of data from multiple channels merging to one; or, replication of data from one channel to many. All this happens in one synchronous step.

Table 1 illustrates a set of primitive channels. The synchronous channel accepts data at the source and dispenses data through the sink as soon as both source and sink are ready. The lossy synchronous channel can always accept data at the source. The data flows from the source to the sink if the sink can accept data at that instance; otherwise, it is lost. The synchronous drain has two source ends; it takes the data on its sources if and only if they are both ready. It acts like a channel synchronizer and does not transfer any data. The FIFO1 channel transfers data from the source to the sink in two transitions, thereby acting like a one-place data storage. A FIFO channel can also be unbounded.

Transitions of constraint automata are labeled with a set of port names and a data constraint. A transition is taken when all of the ports on its label are ready. In that case, the data constraint determines the data flow in a declarative
Table 1. Basic Channels and their constraint automata.

<table>
<thead>
<tr>
<th>Channel Type</th>
<th>Automaton Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Synchronous channel</td>
<td>Data(A) = Data(B)</td>
</tr>
<tr>
<td>FIFO1 channel with variable d</td>
<td>d = Data(A)</td>
</tr>
<tr>
<td>Lossy synchronous channel</td>
<td>Data(A) = Data(B)</td>
</tr>
<tr>
<td>Synchronous drain</td>
<td>Data(B) = d</td>
</tr>
</tbody>
</table>

fashion, e.g., when a synchronous channel fires the data at both ends will be the same. Direction of data flow is understood from the types of channel ends. As in FIFO1, a CA can have variables to temporarily store data values. The initial state of the CA for FIFO1 depends on whether it is initially full or empty.

When channels are composed into a connector, the behavior of the connector is derived compositionally as the product of the CA of its constituent channels. Furthermore, the hiding operator can be applied to create a simple and intuitive CA that accurately describes how the connector works, without exposing the internal ports. Please refer to [5] for a formal definition of product and hiding.

3.1 Integrating Real-Time Actors with Reo

Integrating actors with Reo is complicated by the asynchronous nature of actors: Actors can send messages whenever they have to; therefore, a Reo connector may not block them exogenously. A natural way of solving this issue is to add a FIFO channel as a message buffer at every input port of a Reo connector. The problem is that for model checking, a suitable bound for these FIFOs is necessary. Furthermore, the number of buffers needed quickly blows up the state-space. As a workaround, we suggest using the buffers that already exist in the actors for this purpose. Nevertheless, the upper bound for these buffers need not be increased as discussed below. This approach can be thought of as a low-level optimization of the schedulability check, where we produce a behavior which at a high-level is indistinguishable from adding buffers to the input ports of the connectors. Before explaining the details, we need to restrict the allowable Reo connectors.

A Reo connector may not lose a message. In fact when a message is lost, it can never meet its deadline, and the system will not be schedulable. If we were to allow lossy connectors, one may argue that lost messages can be seen as having met their deadlines; this can be justified by assuming that the Reo connector is in charge and has rightfully decided to lose the message. But this causes a problem if the connector has a buffer to store messages before they are lost (which is the case as explained above). Since we assume that a Reo connector operates in zero time, it may lose any arbitrary number of messages.
in zero time and therefore, we cannot statically compute a bound on the size of this buffer for a schedulable system. This restriction, however, does not greatly reduce the expressiveness of Reo as witnessed by the examples provided in this section and in Section 5. Notice that drain and lossy synchronous channels can still be used.

Another restriction is that only bounded FIFO channels may be used. Therefore, the CA for these connectors is finite-state. Now we explain how to optimize analysis for two patterns of Reo connectors:

- **Single-input, multiple-output (e.g. Fig. 5.a):** Since the output ports are directly connected to a message buffer in an actor, they are always enabled. Therefore, as soon as there is a message on the input port of this connector, it can decide the destination of the message. Since the connector does not lose the message, it may directly go to an actor or it is stored in a FIFO channel. In either case, we do not need an extra buffer at the input port.

- **Multiple-input, single-output (e.g. Fig. 5.b):** In this case, the destination of all messages is the same, namely the actor connected to the output node. Therefore, we can reuse the buffer of this actor to hold also the messages pending at the input ports. To distinguish these messages from the ones actually in the actor's buffer, these pending messages are flagged so that the actor scheduler cannot select them. This flag will be removed from a message whenever the Reo connector decides that this message can actually be delivered to the recipient actor.

As a consequence, the Reo models do not need to include extra buffers at the input, and rather focus on the coordination logic (cf. Fig. 5). Compared to a normal buffer (as in Section 2, disabling a message only delays its execution, whereas its deadline counts since it is generated. Therefore, as before, a queue with more than \( \lceil d_{\max} / b_{\min} \rceil \) messages is not schedulable. Subsection 3.2 describes how we can implement the above solutions in Uppaal. Section 5 introduces more patterns in which such optimizations are possible.

**Client-Server connectors** In our example of client-server we have one client and two servers. The requests and replies between the client and the servers are routed through the connectors shown in Fig. 5. The request sequencer accepts messages from the client through the input port \( I \) and routes them to the servers through the output ports \( O_1 \) and \( O_2 \) in a strict sequence. The reply sequencer accepts messages through input ports \( I_1 \) and \( I_2 \) and routes them back to the client through output port \( O \), in the order in which they were sent. In both connectors we have a circular configuration of FIFO1 channels, this is to produce an alternating behavior of port selection. For the request sequencing we see that one FIFO1 channel is initially full, this causes ports \( \{I, x_0, x_1, x_2, f_1, O_1\} \) to become enabled when a message is put on input port \( I \) and the request flows through output port \( 1 \ O_1 \). Now the FIFO1 channel is full, so for the next request the ports \( \{I, x_0, x_3, x_2, f_1, O_2\} \) are enabled for the next message, causing the data to flow through output port \( O_2 \). Similarly, for the request sequencing
we have that FIFO$_1$ channel is initially full, which forces a strict sequencing on the order in which the replies are put into the buffer of the client. To avoid blocking the input ports $I_1$ or $I_2$, in principle we need to add extra buffers on the input ports; this extra buffer is avoided by reusing the buffer of Client as we explained above. In the next section, we show how to implement this in UPPAAL.

In the sequel, we hide internal ports $\{x_0, x_1, x_2, x_3, f_1, f_2\}$ in the CA models.

### 3.2 Analysis in UPPAAL

To be able to perform analysis in UPPAAL, we need to give a representation of CA in terms of UPPAAL timed automata. We work with the CA representing each connector, i.e., after the product of the CA of the constituent channels has been computed. Furthermore, all internal ports should be hidden. Therefore, we are not concerned with composing two translated CA.

The idea is that synchronization on port names can be translated to channel synchronization in UPPAAL. We can reuse the invoke channel for this purpose. Recall from Section 2 that invoke is used for sending messages. An action on an input (resp. output) port is translated to a ‘receive’ (resp. ‘emit’) on the channel. Variables in CA can be directly translated to variables in UPPAAL, therefore, data constraints can be simply translated to assignments in UPPAAL.

The main challenge is that transitions in CA may require synchronization on multiple ports, whereas in UPPAAL channels provide binary synchronization. To solve this, whenever multiple ports should synchronize, they are put on consecutive transitions separated by committed locations. This produces an equivalent behavior as these transitions are all taken in zero time and without being interleaved with other automata instances. In the following, we show how to implement the optimizations for the two Reo patterns mentioned previously.

- **Single-input, multiple-output**: In this case (e.g., the request sequencer), the message can immediately be processed and the sender will never be
blocked. Therefore, the above translation from CA to timed automata is enough and the CA can directly intermediate between the sender and receiver actors. For example in Fig. 6.(a), the synchronous step on $I$ and $O_1$ is modeled by first reading a request message on $I$ and then writing the message on $O_1$. Similarly, $I$ and $O_2$ are synchronized at the next step.

- **Multiple-input, single output:** As explained in previous subsection, actor buffers need to be extended such that every message has a boolean flag called ‘enabled’. As long as this flag is false, the message will not be selected by the scheduler. The extended `insertInvoke` function (cf. Section 2) assigns variable ‘constraint’ to the ‘enabled’ field corresponding to every incoming message. The variable ‘constraint’ is always set to true, except when a message is sent via a “multiple-input, single-output” connector (cf. Fig. 6.(b)). Via this connector, all messages are directly passed on to the buffer of the single receiver with their ‘enabled’ flag set to false.

Another automaton, shown in Fig. 6.(c), captures the coordination logic, i.e., it has the exact form of the constraint automata for the Reo connector. The second automaton in Fig. 6.(c) is an extension to the scheduler automata which follows the coordination logic to enable messages in the queue. Therefore, these messages are enabled at the moment that is allowed by the CA. In this figure $q[i]$ shows a message at index $i$ of the queue which was sent by $s[i]$. Note that this automaton selects only disabled messages, i.e., it does not consider a message twice. However, as shown in this figure, it does not distinguish between different instances of the same message. Since every message already has a clock assigned to it which keeps track of how long it has been in the queue, we can use that clock to select the oldest message instance. To do so, we need to extend the guard like this:

```plaintext
i < tail && ! enable[i] &&
forall (m : int[0..MAX-1]) (
    enable[m] || m>=tail ||
    q[i] != q[m] || s[i] != s[m] ||
    clk[ca[i]]-clk[ca[m]]>=0
)
```

where $\text{clk}[ca[i]]$ shows the clock assigned to the message at $q[i]$. 

---

Fig. 6. Integrating Constraint Automata into UPPAAL.
Compatibility Check To check the compatibility of actors coordinated using Reo connectors, we need to compose the behavioral interfaces of the actors with the Constraint Automata models of the Reo connectors. This composed automaton will serve as the basis for test case generation. In this composition, we will use the transformed version of the constraint automata into UPPAAL format. However, the coordinate channels need to be converted back to invoke channel so that the behavioral interfaces can communicate with them. Note that converting Constraint Automata to Timed Automata can ideally be automated such that these conversions would be safe from human error.

Fig. 7 shows a new behavioral interface for the client that accommodates a late reply by incorporating the possibility of sending two requests in a row. On the right side, a (simplified) test case is shown that is generated from the composition of behavioral interfaces of one client and two servers connected with the sequencer Reo connectors. Compared to the test case in Section 2, this test case can identify two servers S1 and S2. This test case cannot reach the FAIL verdict. This is because before the client wants to send a third request, the servers will provide the replies.

4 Actors with Communication Delays

In this section, we show how to extend the modeling framework of Section 2 and the corresponding schedulability analysis to take account of communication delays between actors. We assume here that actors communicate directly, i.e., there is no Reo connector.

We assume a fixed delay for communications between every pair of actors, called their distance. This is a reasonable assumption if the communication medium between the actors is fixed for all messages. Therefore, the delays in the whole network can be modeled as a matrix; this matrix will be symmetric if we assume the uplink and downlink connections have the same properties. For example, for the client-server example, we assume the distance 1 between the client and the server (see Fig. 8). The distance of an actor to itself is then zero.

Fig. 8. The distance matrix.
1. The time difference since a message is sent and is executed (at receiver) cannot be smaller than the distance between the sender and the receiver.
2. The deadline associated to each message (specified by the sender) should also include the network delay.
3. The modularity of the analysis techniques should be preserved.

A naive solution to handle network delays is to introduce network buffers, e.g., by adding an extra actor. This actor should delay each message exactly as intended and reduce its remaining deadline correspondingly. This, however, introduces a great overhead in the size of the model: there will be at least a buffer (and its corresponding clocks) between every pair of actors in each direction, i.e., an exponential number of buffers and clocks. Additionally, finding a reasonable upper bound on the size of these buffers is not trivial.

To avoid introducing this overhead, we place the messages directly into the buffer of the receiving actors. To model the distances, the messages in the buffer should be disabled as long as the network delay has not passed (concern 1). As explained in Section 2, $\text{clk}[\text{msg}]$ is reset to zero when $\text{msg}$ is added to the buffer. With the distance matrix available, we can use this guard:

$$\text{distance}[\text{sender}][\text{receiver}] < \text{clk}[\text{msg}]$$

as the enabling condition for each message. Recall that scheduling policies are implemented as guards in the scheduler automata in UPPAAL, which model the selection condition of every message. The above enabling condition can therefore be hard coded into this guard. Thus we avoid extra variables in the buffer representation to capture the enabling conditions of messages, which leads to a very efficient implementation. Additionally, using $\text{clk}[\text{msg}]$ together with the original deadline of the message satisfies the second concern in a straightforward way.

This approach brings about two new concerns:

4. In this approach, the order of messages in a buffer are based on their sending time rather than their arrival time, i.e., when they become enabled.
5. While preserving schedulability, the buffer of every actor needs to be big enough to contain all messages, including disabled messages.

Since messages may arrive from different actors with different distances, multiplexing them into the same buffer should preserve their order of arrival rather than their order of sending. This is important in scheduling strategies that depend on the arrival order of the messages, e.g., FIFO. To address this issue, we need to re-implement such schedulers based on the waiting time of messages after they become enabled, which is equal to $\text{clk}[\text{msg}] - \text{distance}[\text{sender}][\text{receiver}]$; this value should be used when it is not negative.

Finally, we show that the size of buffer for schedulable actors does not need to be increased in presence of network delays. As argued in previous section, disabling a message may only delay its execution, whereas the deadline associated to all messages (disabled or enabled) is still in effect and approaching. Therefore, if there are $n > \lceil d_{\text{max}} / b_{\text{min}} \rceil$ messages in the buffer, one of them inevitably
misses its deadline. This means that individually schedulable actors can still be used provided that the compatibility check is adapted, i.e., modularity is preserved.

4.1 Compatibility Check

Definition 2 defines compatibility as the inclusion of visible traces of the system $S$ in the traces of $B$, where $B$ is the composition of the behavioral interfaces. The actions in these traces are instantaneous communication of messages; however, in presence of network delays, communication is not instantaneous any more. The main challenge here is to bridge the time gap between the traces in $S$ which capture the sending times and the traces in $B$ which reflect the arrival times.

**Definition 3 (Compatibility with delay).** For every trace from $S$, say $\sigma = (m_1, t_1) \ldots (m_i, t_i) \ldots$, which captures the sending time of each message, there should exist a corresponding trace $\sigma' = (m_1, t_1 + x_1) \ldots (m_i, t_i + x_i) \ldots$ in $B$, where $x_i$ is the distance between the sender and receiver of $m_i$ (cf. Fig. 8).

Furthermore, a deadline on the server side (in the behavioral interface) only includes the buffer time and the execution time, whereas a deadline on the client side (in a method) includes also the network delay. In other words, the compatibility check must ensure that the client side deadline is not smaller than the deadline on the server side plus the distance between the actors.

To check compatibility, as explained in Section 2, we generate test cases from the more abstract side, i.e., the composition of the behavioral interfaces $B$. A test-case in the original framework [16] both drives the system under test and monitors it for unexpected behavior. These tasks must be separated now: a test-driver automaton communicates with the system based on the send times (cf. Fig. 9.(a)); a monitor automaton checks whether the arrival time of messages matches the expectations in $B$ (cf. Fig. 9.(b)). The latter is not trivial as the arrival time of a message is when it becomes enabled. Therefore, the scheduler automata must send a signal on a new channel, check, at the actual arrival time of the message, i.e., $clk[msg]$ reaches $distance[sender][receiver]$ (cf. Fig. 9.(c)).

A test-driver is a linear timed automaton generated from a trace taken from $B$. To be able to drive the system under test, the arrival times must be changed to sending times. As a result, we may need to reorder the transitions of the original trace so that the messages are sent in the correct chronological order.
The monitor automaton is obtained in the same way as in Section 2 when no delays are present. However, it does not drive the system behavior any more. Instead it uses the check channel to see if an actor in the system could receive a message outside the expected time as specified in its behavioral interface. Fig. 9(b) considers the client/server model in Section 2 where two consecutive request messages are disallowed.

5 Discussion and Future Work

We extended our previous work on schedulability analysis of real-time actors to consider complex networks of actors. On one hand, the coordination language Reo is applied. Reo can be used to take better advantage of off-the-shelf components, where in our case components are modeled as actors. We showed with our simple example that with the help of Reo we can combine actors in such a way that their combination becomes schedulable; in addition, more complicated systems can be built. On the other hand, we showed how to consider communication delays between actors. This is especially important when actors are to be deployed on remote machines.

In an ideal situation, Reo connectors can carry timing information and as such also include the network delays. However, as already mentioned, there is currently no fully satisfactory real-time extension of Reo. As a result, we continue with the assumption that the coordination in Reo connectors happens in negligible time (as in Section 3). Furthermore, we assume that Reo connectors are local to actors. Therefore, the use of a distance matrix as introduced in Section 4 is orthogonal to using Reo. This means that one can directly combine the techniques in the previous two sections to analyze coordinated networks of actors in presence of delays.

In Fig. 10 we illustrate this implementation graphically for our running example. The request and reply sequencing connectors are local to the Client actor. The real delay happens in the network cloud (formally modeled in the distance matrix). By assuming a fixed delay between every pair of actors, we can essentially look at the network as a black-box, i.e. we don’t need to know any details about the network, only how long it takes to send messages through the network.

For checking compatibility, we need to generate the separate test-driver and monitor automata because of the delay in the network. Nevertheless, the test cases should be generated from the composition of the behavioral interfaces and the constraint automata models of the Reo connectors, as depicted in Section 3.
Reo Patterns. In this paper, we considered only two patterns of Reo connectors, i.e., single input or single output. Although this may seem a strict restriction on use of Reo, many useful connectors can still be used. Another example of such connectors is shown in Fig. 11.(a). In this example, the client actor requires two services $m_1$ and $m_2$ (say ‘BookFlight’ and ‘BookHotel’) but there is no server actor that can provide both. The connectors in this figure can be used to connect such a client to two servers each providing one of these services. In this connector filter channels are used which may pass the incoming data only if it matches the pattern provided and thus e.g. distinguishing $m_1$ and $m_2$. The replies from the two servers can be simply merged using a merger as shown in Fig. 11.(b).

Although applying a multiple input multiple output connector may in general require an extra buffer at its input, this can be avoided again in several kinds of connectors, which need to be considered individually. Another example where we can optimize the implementation is a barrier synchronizer, shown in Fig. 11.(c). A barrier synchronizer delays the messages from the fast client actors until all inputs are ready and only then forwards them to their destinations. In this connector, the destination actor for each input port is statically known; therefore, the buffer of that actor can be used to store messages on the respective input port.

References


A Formal Model of Object Mobility in Resource-Restricted Deployment Scenarios *

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Abstract. Software today is often developed for deployment on different architectures, ranging from sequential machines via multicore and distributed architectures to the cloud. In order to apply formal methods, models of such systems must be able to capture different deployment scenarios. For this purpose, it is desirable to express aspects of low-level deployment at the abstraction level of the modeling language. This paper considers formal executable models of concurrent objects executing on deployment components, which reflect the execution capacity of sets of objects between observable points in time. We show how to model and simulate the behavior of objects moving between components, by means of relocation strategies expressed in the modeling language. A running example is used to demonstrate how activity on deployment components causes congestion and how object relocation can alleviate this congestion. We simulate the behavior of models which vary in the execution capacity of deployment components and in object relocation strategies.

1 Introduction

Software is increasingly often developed as a range of systems. Different versions of a software may provide different functionality and advanced features, depending on target users. In addition to such functional variability, software systems need to adapt to different deployment scenarios. For example, operating systems adapt to specific hardware and even to different numbers of available cores; virtualized applications are deployed on a varying number of (virtual) servers; and services on the cloud may need to adapt dynamically to the underlying cloud infrastructure. This kind of adaptability raises new challenges for the modeling and analysis of component-based applications [33]. To apply formal methods to such applications, it is interesting to lift aspects of low-level deployment concerns to the abstraction level of the modeling language. In this paper we propose abstract performance analysis for formal object-oriented models, in which objects may migrate between deployment components that are parametric in the amount of concurrent processing resources they provide to their objects.

The work presented in this paper uses ABS [20], a modeling language for distributed concurrent objects which communicate by asynchronous method calls.

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ABS is an executable language, but still allows abstractions (i.e., functions and abstract data types can be used to specify internal, sequential computations). The concurrent object semantics of ABS are similar to Actors [1] and Erlang processes [5]: ABS objects are inherently concurrent, with at most one process active per object. Concurrent objects and agents have attracted attention as an alternative to multi-thread concurrency in object-orientation (e.g., [9]), and been integrated with, e.g., Java [30,32] and Scala [15]. ABS proposes cooperative scheduling of processes inside concurrent objects, which eliminates some common programming errors (specifically, race conditions are much harder to introduce inadvertently) and enables compositional verification of ABS models [2,12].

In order to capture deployment scenarios for ABS models, previous work by the authors proposes an extension of the ABS language with deployment components which are parametric in the amount of concurrent activity they allow within a time interval [22]. This allows us to analyze how the amount of concurrent execution resources allocated to a deployment component influences the performance of objects deployed on the component. For this purpose, we work with a notion of timed concurrent objects [8], extended to capture parametric concurrent activities between observable points in time. To validate and compare the concurrent behavior of models under restricted concurrency assumptions, the timed operational semantics of our ABS extension, for brevity presented in an SOS style [29] in this paper, is expressed in rewriting logic [25], which enables the use of Maude [11] for as a simulation and analysis tool for ABS models.

The contribution of this paper is a formalization of object mobility in resource-restricted deployment scenarios. We extend ABS with a goto statement which allows objects to move between deployment components, show how this extension naturally integrates in the ABS semantics in an elegant and simple way, and how it allows load balancing strategies to be expressed and executed in parallel with the functional parts of the model. This approach complements the work presented in [21], which formalizes load balancing by resource reallocation. We demonstrate the use of simulation techniques to analyze the resource usage of distributed system models in ABS in order to compare the behavior of models ranging over resources and load-balancing strategies. This enables designers to anticipate behavior of distributed systems at an early stage in the design process.

**Paper Overview.** Section 2 introduces the language via a running example. Section 3 presents the syntax and semantics of a timed concurrent object modeling language with deployment components. Section 4 shows how we can use our interpreter to simulate the behavior of our example ranging over deployment scenarios. Section 5 talks about load balancing strategies, Section 6 discusses related work, and Section 7 concludes the paper.

## 2 Motivating Example

Let us consider services for building software using the service-oriented design paradigm; e.g., as found in web shops, social networks, browsers applications. The ABS model of such a service is given in Fig. 1 (Sec. 3 contains a detailed
interface Agent {Session getSession(); Unit free(Session session);}  
interface Session {Bool request(Int cost);}  

class SessionImp(Agent agent) implements Session {
    Bool request(Int cost) {
        Time start = now;
        Int remCost = cost;
        while (remCost > 0) {remCost = remCost - 1;}
        agent.free(this);
        return (2 * (now - start) < cost);}
}

class AgentImp() implements Agent {
    Set<Session> sessions = Empty;
    Unit free(Session session) {sessions = add(session, sessions);}
    Session getSession() {Session session;
        if (!isEmpty(sessions)) {session = new SessionImp(this);}
        else {session = choose(sessions);
            sessions = remove(session, sessions);}
        return session;}
}  

// Main block:
DC server = component(20); Agent a = new AgentImp() in server;

Fig. 1. A web application model in ABS.

An explanation of the language syntax). In our example, clients use the service by first calling the getSession method of an Agent object. An Agent hands out Session objects from a dynamically growing pool. Clients then call the request method of their Session instance. After completing the request, the session object is returned to the agent’s pool. For simplicity, we abstract from the specific functionality of our service and let the request method of a session have a certain computation cost, given as the actual parameter to the method. This cost reflects the computation cost of the service given in terms of its input size; e.g., manipulating photos or recursing through an address list (where remCost reflects the remaining cost of the computation of the request method). A request is successful if it can be handled within a certain amount of time, depending on its input. This is captured by the return statement of the method, which relates the execution time of a method activation (i.e., the difference between the current time now and the start time of the request) to the input cost of the method call.

In the Agent class, the attribute sessions stores a set of Session objects. (ABS has a datatype for sets, with operations isEmpty to check for the empty set, denoted Empty, choose to select an element of a non-empty set, and the usual remove and add operators.) When a customer requests a Session, the Agent takes a session from the available sessions if possible, otherwise it creates a new session. The method free inserts the session in the available sessions of the Agent, and is called by the session itself upon completion of a request. This model captures the architecture and control flow of a service oriented application, while abstracting from many details (such as thread pools, data models, sessions spanning multiple requests etc.) which can be added to the model if needed.
The main block of the model specifies the initial state for model execution as a deployment scenario in which an agent object is deployed on a deployment component server (of the predefined type DC), which will also contain the Session objects. The parameter to the server specifies its execution capacity in terms of abstract concurrent resources, which reflect the amount of potential abstract execution cycles available to the objects deployed on the server between observable points in time. The agent creates concurrently executing Session objects on the same server as needed. It is easy to see that heavy client traffic may lead to congestion on the server, which may in turn cause a lot of unsuccessful requests to the service.

3 Models of Deployed Concurrent Objects in ABS

ABS is an abstract behavioral specification language for distributed concurrent objects. This section briefly introduces ABS (for further details see, e.g., [20]). In ABS, objects conceptually have dedicated processors and live on deployment components which constrain concurrent execution capabilities. Objects are dynamically created instances of classes, with attributes initialized to default type-correct values. An optional init method may be used to redefine attributes. Objects are typed by interface and communicate by asynchronous method calls, spawning concurrent activities in the called object. Active behavior, specified by an optional run method, is interleaved with passive behavior, triggered by such asynchronous method calls. Thus, an object has a set of processes to be executed, which stem from method activations. Among these, at most one process is active. The others are suspended on a queue. Process scheduling is by default non-deterministic, but controlled by processor release points in a cooperative way. ABS is strongly typed: for well-typed programs, invoked methods are supported by the called object (when not null), and formal and actual parameters match. We assume that programs are well-typed.

Deployment components restrict the inherent concurrency of objects in ABS by mapping the logical concurrency to a model of physical computing resources. In ABS, objects exist in the context of a deployment component with a given amount of resources. Deployment components are first-class citizens of ABS, of type DC. Deployment components abstract from the number and speed of the physical processors available to the component by a notion of concurrent resource [22]: A simple time model defines the points in time when the executing system is observable. Concurrent resources can be consumed in parallel or in sequential order, which reflects the number of processors and their speeds relative to the intervals between observable points in time. Objects deployed on a component may consume resources within a time interval until the component runs out of resources or the objects are otherwise blocked. This way, the logical concurrency model of the concurrent objects is controlled by their associated deployment component. A deployment component is parametric in the computational resources it offers to a group of dynamically created objects, which makes it easy to configure deployment scenarios varying in their concurrent resources.
Syntactic categories. Definitions.

$C, I, m$ in Names

$g$ in Guard

$s$ in Stmt

$x$ in Var

$e$ in Expr

$b$ in BoolExpr

$r$ in Resource

$\text{Syntactic categories. Definitions.}$

\begin{align*}
C, I, m & \text{ in Names} & IF & := \text{ interface } I \{ \exists g \} \\
g & \text{ in Guard} & CL & := \text{ class } C \{ \exists x \} \{ \text{ implements } T \} \{ \exists \} M \\
s & \text{ in Stmt} & Sg & := I \{ \exists T \} x \\
x & \text{ in Var} & M & := Sg \{ \exists x \} s \\
e & \text{ in Expr} & g & := b \mid x? \mid g \land g \\
b & \text{ in BoolExpr} & s & := s; s \mid x = rhs \mid \text{suspend} \mid \text{await} \mid g \mid \text{skip} \mid \text{goto}(e) \\
r & \text{ in Resource} & e & := x \mid b \mid r \mid \text{this} \mid \text{now} \mid \text{thiscomp} \mid \text{available} \mid \text{load}(e) \\
rhs & := e | cm \mid \text{new } C(\tau) \mid \text{in } e \mid \text{component } (e) \\
cm & := [e]!m(\tau) | [e.]m(\tau) | x \cdot \text{get} \\
\end{align*}

Fig. 2. ABS syntax. Terms such as $\tau$ and $\pi$ denote lists over the corresponding syntactic categories, square brackets $[ ]$ denote optional elements.

Figure 2 gives the syntax of timed ABS with deployment components. A program consists of interface and class definitions and a main block to configure the initial state. $IF$ defines an interface with name $I$ and method signatures $Sg$. A class implements a set $T$ of interfaces, which specify types for its instances. $CL$ defines a class with name $C$, interfaces $T$, class parameters and state variables $x$ (of type $I$), and methods $M$. (The attributes of the class are both its parameters and declared fields.) A method signature $Sg$ declares the return type $I$ of a method with name $m$ and formal parameters $\tau$ of types $T$. $M$ defines a method with signature $Sg$, a list of local variable declarations $\pi$ of types $T$, and a statement $s$.

Statements. Assignment $x = rhs$, sequential composition $s_1; s_2$, skip, if, while, and return $e$ are standard. The statement goto$(e)$ moves the object to deployment component $e$. The statement suspend unconditionally releases the processor by suspending the active process. The guard $g$ controls processor release in statements await $g$, and consists of Boolean expressions $b$ over attributes and return tests $x$? (see below). If $g$ evaluates to false, the current process is suspended. In this case, any enabled process from the pool of suspended processes may be activated. The scheduling of processes is cooperative in the sense that processes explicitly yield control and execution in one process may enable the further execution in another.

Expressions $rhs$ include pure expressions $e$, communications $cm$, and the creation of deployment components and objects. The expression component$(e)$ creates a component with $e$ concurrent resources. Resources are modeled by a type Resource which extends the natural numbers with an “unlimited resource” $\omega$. The set of concurrent objects deployed on a component, representing the logically concurrent activities, may grow dynamically. Object creation new $C(\tau)$ in $e$ has an optional clause to specify the targeted deployment component: here the $C$ object is to be deployed on component $e$. (If the target component is omitted, the new object will be deployed on the same component as its parent. The behavior of ABS models without deployment restrictions on their functional behavior is captured by a main deployment component with $\omega$ resources.)
Pure expressions \( e \) are variables \( x \), Boolean expressions \( b \), resources \( r \), \texttt{this} (the object’s identifier) and \texttt{thiscomp} (the object’s current deployment component), and \texttt{now}, which returns the current time. Timed ABS uses an implicit time model [8], comparable to a system clock which updates every \( n \) milliseconds (representing a time interval). Time values are totally ordered by the less-than operator; comparing two time values results in a Boolean value suitable for guards in \texttt{await} statements. From an object’s local perspective, the passage of time is indirectly observable via \texttt{await} statements. Time advances when no other activity may occur. This model of time is used to handle the amount of concurrent activity allowed within a time interval in order to model resource constraints for different deployment scenarios. The resources available to objects on the current deployment component are given by \texttt{available}, and the average load on the component for the last \( e \) time intervals by \texttt{load}(e). (The full language includes a functional expression language with standard operators for data types such as strings, integers, lists, sets, maps, and tuples. These are omitted here, and explained when used in the examples.)

Communications \texttt{cm} are based on asynchronous method calls. After making an asynchronous call \( x = e!m(e) \), the caller may proceed without waiting for the method reply. Here \( x \) is a \textit{future variable}, which refers to a return value which may still need to be computed. Two operations on future variables control synchronization in ABS [20]. First, the guard \texttt{await} \( x? \) suspends the active process until a return to the call associated with \( x \) has arrived. This suspends execution of the process, but allows other processes to run. Second, the return value is retrieved by the expression \( x\texttt{.get} \), which \textit{blocks} all execution in the object until the return value is available. Two commonly used communication patterns are now explained; the statement sequence \( x = e!m(e); \ y = x\texttt{.get} \) encodes a blocking call, conveniently abbreviated \( y = e.m(e) \) (often referred to as a synchronous call), whereas the statement sequence \( x = e!m(e); \ \texttt{await} x?; \ y = x\texttt{.get} \) encodes a non-blocking, preemptable call.

3.1 Operational Semantics

The semantics of timed ABS with deployment components is given as a transition system in an SOS style [29]. Transition rules apply to subsets of configurations. For simplicity, configurations can be reordered to match the left hand side of rules (i.e., matching is modulo associativity and commutativity as in rewriting logic [11]). A system run is a possibly nonterminating sequence of rule applications. Auxiliary functions are evaluated between the application of transition rules.

Timed configurations \( tcn \) are given in Fig. 3 and consist of one global clock and an untimed configuration \( cn \); i.e., a set of deployment components, objects, invocation messages, and futures. Let whitespace denote the associative and commutative union operator on configurations and \( \varepsilon \) the empty configuration. Timed configurations live inside curly brackets; in the term \( \{x\} \), the variable \( x \) captures the entire configuration. In the (global) clock \( cl(t) \), \( t \) is the current time. In a deployment component \( dc(n, r, u, h) \), \( n \) is the identifier of the component, \( r \) the (non-negative) number of available computing resources, \( u \) the maximum
The syntax for timed runtime configurations.

\[
\begin{align*}
\text{cn} & ::= \epsilon | \text{comp} | \text{object} | \text{msg} \\
& \quad | \text{future} | \text{cn} \ cn \\
\text{object} & ::= \text{ob}(o, \sigma, p, q) \\
\text{msg} & ::= \cdot \text{t} \ \text{future} | \text{cn} \ cn \\
\text{future} & ::= \text{fut}(f, v) \\
\text{comp} & ::= \text{dc}(n, r, u, h) \\
\text{clock} & ::= \text{cl}(t) \\
\sigma & ::= x \mapsto v \ | \sigma[x \mapsto v] \ | \sigma, \sigma \\
n & ::= a \ | f \ | \text{null} \ | b \ | t
\end{align*}
\]

Fig. 3. The syntax for timed runtime configurations.

The rewrite rules of the operational semantics transform state configurations into new configurations, and are given in Fig. 4 and Fig. 5.
\( \text{(Assign1)} \)

\[
\begin{align*}
x & \in \text{dom}(l) \quad \implies \quad v = [e]^t_{(a; s)} \\
\text{a(thiscomp)} & = n \quad r > 0 \\
\text{ob}(o, a, \{l[x := e; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l[x := e; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{(Assign2)} \)

\[
\begin{align*}
x & \in \text{dom}(a) \quad \implies \quad v = [e]^t_{(a; s)} \\
\text{a(thiscomp)} & = n \quad r > 0 \\
\text{ob}(o, a, \{l[x := e; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l[x := e; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{(Async-Call)} \)

\[
\begin{align*}
\text{ob}(o, a, \{l'[x := e; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l'[x := e; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{(Async-Call2)} \)

\[
\begin{align*}
\text{ob}(o, a, \{l[x := e; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l[x := e; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{(Await1)} \)

\[
\begin{align*}
\text{ob}(o, a, \{l[\text{await } g; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l[\text{await } g; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{(Await2)} \)

\[
\begin{align*}
\text{ob}(o, a, \{l[\text{release; await } g; s] \}, q) & \text{ dc}(n, r, u, \overline{h}) \\
& \implies \text{ob}(o, a, \{l[\text{release; await } g; s] \}, q) \text{ dc}(n, r - 1, u, \overline{h})
\end{align*}
\]

\( \text{Fig. 4. Operational semantics of resource-restricted timed ABS (1).} \)

If \( \sigma \) is a mapping, denote by \( \text{dom}(\sigma) \) its domain; by \( \sigma(x) \) the value bound to \( x \) in \( \sigma \) (assuming \( x \in \text{dom}(\sigma) \)); by \( \sigma[x \mapsto v] \) the extension of \( \sigma \) which binds \( x \) to \( v \) (and \( \sigma[x \mapsto v](x') = \sigma(x') \) if \( x \neq x' \)); and by \( \sigma_1 \circ \sigma_2 \) the composed mapping such that \( \sigma_1 \circ \sigma_2(x) = \sigma_2(x) \) if \( x \in \text{dom}(\sigma_2) \), \( \sigma_1 \circ \sigma_2(x) = a(x) \) otherwise. For simplicity, classes are not represented explicitly in the semantics, but may be seen as static tables. Assume given functions \( \text{bind}(o, f, m, \overline{\tau}, C) \) which returns a process resulting from the method activation of \( m \) in a class \( C \) with actual parameters \( \overline{\tau} \), callee \( o \) and associated future \( f \); \( \text{init}(C) \) which returns a process initializing instances of class \( C \); and \( \text{atts}(C, \overline{\tau}, o, n) \) which returns the initial state of an instance of class \( C \) with class parameters \( \overline{\tau} \), identity \( o \), and deployment component \( n \). The predicate \( \text{fresh}(n) \) asserts that a name \( n \) is globally unique (where \( n \) may be an identifier for an object or a future). Let ‘idle’ represent any process \( \{l | s \} \) where \( s \) is an empty statement list. We define different assignment rules for side effect free expressions (Assign1 and Assign2), object
creation (New-Object1 and New-Object2), method calls (Async-Call), and future dereferencing (Read-Fut).

Rule Skip consumes a skip in the active process and a resource in the object’s deployment component. Here and in the sequel, the variable s will match any (possibly empty) statement list, the object’s deployment component is given by a(thiscomp), and r > 0 asserts that the deployment component has available resources. Rules Assign1 and Assign2 assign the value of expression c to a variable x in the local variables l or in the fields a, respectively, consuming a resource in the deployment component of the object. Rules Cond1 and Cond2 branch the execution depending on the value obtained by evaluating the expression c. (We omit the rule for while, which unfolds the while loop using an if-expression.)

Process Suspension and Activation. Three operations are used to manipulate a process queue q: enqueue(p, q) adds a process p to q, q \ p removes the process p from q, and select(q, a, cn, t) selects a process from q (if q is empty, this is the idle process or no process is ready). The actual definitions of these operations are left unspecified; different definitions correspond to different scheduling policies for processes (e.g., EDF, FIFO). Let \emptyset denote the empty queue. Rule Release suspends the active process to the process queue, leaving the active process idle. Rule Await1 consumes the await statement if the guard evaluates to true in the current state of the object, rule Await2 adds a release statement in order to suspend the process if the guard evaluates to false. Rule Activate selects a process from the process queue for execution if this process is ready to execute, i.e., if it would not directly be resuspended or block the processor.

Fig. 5. Operational semantics of resource-restricted timed ABS (2).

\[
\begin{align*}
\text{(release)} & \quad \text{ob}(o, a, \{l|\text{release}; s\}, q) \\
& \quad \quad \rightarrow \text{ob}(o, a, \text{idle}, \\
& \quad \quad \quad \text{enqueue(}\{l|s\}, q)) \\
\text{(Read-Fut)} & \quad v \neq \perp, f = [e]^{\varepsilon}_{l(t)}, \ a(\text{thiscomp}) = n, r > 0 \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x := e; s\}, q) \ f(u, v) \ cc(n, r, u, h) \ cl(t) \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x := v; s\}, q) \ f(u, v) \ dc(n, r - 1, u, h) \ cl(t) \\
\text{(Go-To)} & \quad a’ = a(\text{thiscomp}) \rightarrow [e]^{\varepsilon}_{l(t)} \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|\text{goto}; e(s); s\}, q) \ cl(t) \\
\text{(Activate)} & \quad p = \text{select}(q, a, cn, t) \\
& \quad \quad \rightarrow \{\text{ob}(o, a, \text{idle}, q) \ cl(t) \ cn\} \\
& \quad \quad \rightarrow \{\text{ob}(o, a, p, q(p)) \ cl(t) \ cn\} \\
\text{(progress)} & \quad \text{canAdv(cn, t)} \\
& \quad \quad \rightarrow \{\text{cn} \ cl(t)\} \\
& \quad \quad \rightarrow \{\text{Adv(cn)} \ cl(t + 1)\} \\
\text{(New-Object1)} & \quad a(\text{thiscomp}) = a’(\text{thiscomp}) \quad p = \text{init}(B) \\
& \quad \quad \rightarrow \text{fresh}(a’’) \ a’’ = \text{atts}(B, [e’’]^{\varepsilon}_{l(t)}, t, a’’, n) \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x := \text{new} B(e); s\}, q) \ cl(t) \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x := a’’; s\}, q) \ ob(a’, a’’, p, \emptyset) \ cl(t) \\
\text{(New-Object2)} & \quad a(\text{thiscomp}) = [e’’]^{\varepsilon}_{l(t), t} \ p = \text{init}(B) \\
& \quad \quad \rightarrow \text{fresh}(a’’) \ a’’ = \text{atts}(B, [?]^{\varepsilon}_{l(t), t}, l’’, [e’’]^{\varepsilon}_{l(t), t}) \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x = \text{new} B(e); s\}, q) \ cl(t) \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x := a’’; s\}, q) \ ob(a’, a’’, p, \emptyset) \ cl(t) \\
\text{(New-Component)} & \quad \text{fresh}(n’’) \ r’ = [e’’]^{\varepsilon}_{l(t), t} \ r \geq r’ \\
& \quad \quad \rightarrow \text{ob}(o, a, \{l|x = \text{component}(e); s\}, q) \ cl(t) \\
& \quad \quad \quad dc(n, r, u, h) \\
& \quad \quad \quad \rightarrow \text{ob}(o, a, \{l|x := n’’; s\}, q) \ cl(t) \\
& \quad \quad \quad \quad dc(n, r - r’, u, h) \ dc(n’, r’, 0, \varepsilon)
\end{align*}
\]
canAdv(cn′, t) = true  cn′ contains no objects or messages
canAdv(msg cn, t) = false  messages are instantaneous
canAdv(ob(o, a, p, q) dc(n, 0, u, H) cn, t)  no more resources
= canAdv(dc(n, 0, u, H) cn, t) ∧ a(thiscomp) = n
canAdv(ob(o, a, {l | x = f; get; s}, q) fut(f, ⊥) cn, t)  o is blocked and
= canAdv(fut(f, ⊥) cn, t)  no value is available
canAdv(ob(o, a, idle, q) cn, t)  no ready processes
= canAdv(cn, t) ∧ select(q, a, cn, t) == idle
canAdv(ob(o, a, p, q) cn, t) = false  otherwise
Adv(dc(n, r, u, H) cn) = dc(n, u, H o u - r) Adv(cn)
Adv(cn) = cn  otherwise

Fig. 6. Auxiliary functions controlling time advance. Here, msg denotes a message and cn′ ranges over message- and object-free configurations.

Communication, Object and Component Creation, and Object Mobility. Rule Async-Call sends an invocation message to o′ with the unique identity f (by the condition fresh(f)) of a new future, the method name m, and actual parameters v. Note that the return value of the new future f is undefined (i.e., ⊥). This operation consumes a resource. Rule Bind-Mtd consumes an invocation method and places the process corresponding to the method activation in the process queue of the callee. The reserved local variable ‘destiny’ is used to store the identity of the future associated with the call. Rule Return places the return value into the call’s associated future. This operation consumes a resource. Rule Read-Fut dereferences the future f if it has a value. This operation consumes a resource. Note that if this attribute is ⊥ the reduction in this object is blocked.

The rules New-Object1 and New-Object2 create a new object with a unique identifier o′. The object’s fields are given default values by atts(B, v, o′, n), extended with the actual values v for the class parameters, o′ for this, and n for thiscomp. In order to instantiate the remaining attributes, the process p is loaded (we assume that this process reduces to idle if init(B) is unspecified in the class definition, and that it asynchronously calls run if the latter is specified). This operation reduces to an assignment which consumes a resource. Note that in New-Object1, the new object inherits the deployment component of its creator, while in New-Object2 the new object is in an explicitly named deployment component. Rule New-Component creates a new component with a fresh identifier n′ and transfers the specified resources r′ from the component n of the creator to n′. The rule can only be applied if n has sufficient resources, and consumes r′ resources from n. Object relocation is captured by the rule Go-To which reassigns the object variable thiscomp to the value of expression e, given as a parameter in the goto statement.

Time advance is controlled by the rule Progress, which updates the global clock. We capture a run-to-completion semantics for concurrent execution within the resource bounds of deployment components: all objects must finish their
actions as soon as possible if resources are available. In order to reflect timed concurrent execution with an interleaving semantics, time cannot advance freely. We follow the approach of Real-Time Maude [27], and use a predicate canAdv to specify when time can advance and a function Adv to specify the effect of advancing time on the configuration. The predicate canAdv ranges over configurations and time (see Fig. 6) and can be explained as follows:

- For simplicity, we assume that invocation messages do not take time. Therefore, time may not advance when a message is on its way.
- Time may not advance if any deployment component has remaining resources and any of the component’s objects \( o \) may execute. There are three cases:
  1. the active process in \( o \) is blocked on a value that has become available,
  2. the active process in \( o \) is idle, but a suspended process can be activated.
  3. the active process in \( o \) is not blocked.
- If all deployment components have run out of resources or none of their objects can proceed, then time can advance.

In summary: Time advances if and only if there can be no activity in any object, and no invocation messages are in transit. Once time has advanced, the deployment components get their resources refreshed for the next time interval by the function Adv, which updates a configuration by resetting the free resources of each deployment component to the specified limit and extending the load history of the components with the unused resources of the current time interval. For simplicity time here advances by a single unit.

The operational semantics presented in this section can be directly represented in rewriting logic [25], which allows models to be analyzed using the rewrite tool Maude [11] as an interpreter for the semantics of ABS.
interface Agent {Session getsession(); Unit free(Session session);}
interface Session {Bool request(Int cost); Unit moveto(DC server);}

class SessionImp(Agent agent) implements Session {
    Time start = now;
    Bool request(Int cost){ ... } // As before
    Unit moveto(DC server){if (thiscomp != server){goto(server);}}}

class AgentImp(DC backupserver) implements Agent {
    Set<Session> sessions = Empty;
    Unit free(Session session){ ... } // As before
    Session getsession() { Session session;
        if (isEmpty(sessions)){session = new SessionImp(this);}
        else {session = choose(sessions);
            sessions = remove(session,sessions);
        if ((2*load(0)) > available){ // Move session to backup server
            session.moveto(backupserver);}
        else {session.moveto(thiscomp); // Use original server
            return session;}}}

Fig. 8. An agent which performs load-balancing for the web application.

4 Comparing Resource-Restricted Behaviors

In order to investigate the effects of specific deployment scenarios on the timing behavior of timed software models, we use Maude as an interpreter for the operational semantics to simulate and test ABS models. The test purpose for these scenarios is to reach a conclusion on whether redeployment on a different configuration leads to an observable difference in timing behavior. We compare the behavior of ABS models with the same functional behavior and input when the models are deployed on components with different amounts of resources.

The approach is illustrated by extending the example of Section 2 with a deployment component and a workload scenario. We included a deployment component Server to hold the Agent and Session objects, which were executed in parallel with simulated clients that provide the workload scenario. We defined a Client class (code not shown) which periodically calls request with cost c every t time intervals. The client object(s) model the expected usage scenario and run with unlimited resources.

Figure 7 (left) shows simulations results using one light client which makes 25 requests of cost 20, with one request per time interval, and one heavy client which makes 25 requests of cost 80, with one request every two time intervals. We run the clients concurrently and monitor the behavior of the model for 100 time intervals. The results show that as the resources available to the server are reduced, the quality of service of the web application goes down due to congestion. We see that all requests can be answered with 35 concurrent resources, but only 50% of the requests are answered within the allotted time. Figure 7 (right) shows the load of the server over time; we see that there are no available resources until time 83, and that the activity on the server ends at time 85. In this case, although all requests were answered, half of the requests did not complete in time due to congestion on the server’s deployment component.
interface Agent {
  Session getsession();
  Unit free(Session session);}

interface Session {
  Bool request(Int cost);
  Unit monitor();}

class SessionImp(Agent agent, Time limit, DC backupserver)
  implements Session {
    Time start = now;
    Bool active = False;
    DC origserver = thiscomp;

    this!monitor(); // Asynchronous call to the monitor method
    while (remCost > 0) { remCost = remCost - 1; suspend; }

    active = False;
    agent!free(this);

    return (now - start <= cost);
  }

  Unit monitor() {
    await (now > (start + limit));
    if (active) goto (backupserver);
    if (!active) {
      if (thiscomp != origserver) goto (origserver);}
  }

class AgentImp(Time limit, DC backupserver) implements Agent {
  // Same as in Fig. 1 but creating Session objects with backupserver
}

Fig. 9. Self-monitoring session objects for the web application.

5 Load Balancing Strategies

ABS models may be augmented with load balancing strategies with the aim of decreasing congestion and thus improving the quality of service compared to models with static deployment scenarios. Load balancing strategies may be expressed in ABS using the resource-related language constructs available, load, and goto.

We illustrate how ABS models may be augmented with load balancing strategies using the running example of a web application, and compare the results of load balancing to the base results presented in Section 4. We explore and model two different load balancing strategies; (1) a load-balancing agent which moves sessions to a backup server when the load on the main server is above a given threshold, and (2) self-monitoring sessions which move themselves to the backup server after a certain time. Both of these deployment scenarios are simulated with the same user scenario as in Section 4. Other, more elaborate load-balancing strategies may be modeled possible in the same style.

Figure 8 shows the ABS model of a load-balancing agent which moves sessions to a backup server when the load on the main server is above 50% of the available resources. This is a simple load balancing strategy that tries to minimize the amount of work done on the second server, while keeping quality of service acceptable. Figure 9 shows an ABS model of self-monitoring session objects which move themselves to the backup server if the execution of the current request takes more than a given amount of time (limit). This models a scenario where the primary server processes as many requests as possible while moving long-running requests to the backup server.

Simulations of Load-Balancing Deployment Scenarios. For the simulations of the running example augmented with load balancing strategies, we added a second
deployment component to the initial configurations of Section 4, and let both deployment components have the same capacity. Figure 10 shows the time progression of the load on the two deployment components using the load-balancing agent strategy (left) and the self-monitoring session strategy (right), with both deployment components running with 35 resources. For the load-balancing agent, the servers are active until time 74, with 98% response success rate. For the self-monitoring session strategy, the servers are active until time 54, which means that with 2 servers the web application responds to the 50 request within 54 time intervals, where 94% of the responses are successful.

Figure 11 summarizes all three scenarios (single server, smart agent, and self-balancing sessions) with deployment component capacities ranging from 10 to 55. It can be seen that the load balancing strategies outperform the single server in all cases (as they should, since these scenarios have twice the total number of resources). The simulations show that except in the most constrained scenarios, the self-balancing strategy outperforms the smart agent for our example model.

6 Related Work

Asynchronously communicating software units, known from Actors and Erlang, are interesting due to their inherent compositionality. Concurrent objects with asynchronous method calls and futures combine asynchronous communication with object orientation [2, 9, 30, 32]. In these models, each software unit is also a unit of concurrency. There is a vast literature on formal models of mobility, based on, e.g., agents, ambient calculi, and process algebras, which is typically concerned with maintaining correct interactions between the moving entities with respect to, e.g., security, link failure, or location failure. For non-functional properties, access to shared resources have been studied through type and effect systems (e.g., [17, 18]), QoS-aware processes proposed for negotiating contracts [26], and space control achieved by typing for space-aware processes [7]. Closer to our work, timed synchronous CCS-style processes can be compared

Fig. 10. Left: Two servers with 35 resources per time interval using the load-balancing agent strategy. Right: The same servers using the self-monitoring sessions strategy.
Fig. 11. Comparison of the number of successful responses in 100 time intervals with a total of 50 request (50% low and 50 % high cost).

for speed using faster-than bisimulation [24], albeit without notions of mobility or location. We are not aware of other formal models connecting execution capacities to locations as in the deployment components studied in our paper.

This paper is part of ongoing work on resource-restricted execution contexts for concurrent objects [4, 21, 22]. Whereas [4] considers memory usage, deployment components with parametric concurrent resources were introduced in [22], extending work on a timed rewriting logic semantics for Creol [8]. A follow-up paper considers resources as first-class citizens of the language, formalizing the semantics of ABS with resource reallocation in rewriting logic [21]. In contrast, the present paper considers object mobility using a goto statement to allow an object to move to another deployment component, formalized in a more abstract SOS semantics. Relocation is possible due to the inherent compositionality of concurrent objects [12]: processes are encapsulated inside objects and the state of other objects can only be accessed through asynchronous method calls. This way the object is in control of its own location, which fits with the encapsulation of both state and control in the concurrent object model. Resource reallocation and object mobility are in a sense complementary means to achieve load balancing: both have applications where they seem most natural.

Techniques and methodologies for predictions or analysis of non-functional properties are based on either measurement or modeling. Measurement-based approaches apply to existing implementations, using dedicated profiling or tracing tools like, e.g., JMeter or LoadRunner. Model-based approaches allow abstraction from specific system details, but depend on parameters provided by domain experts [13]. A survey of model-based performance analysis techniques is given in [6]. Formal approaches using Petri Nets, game theory, and timed automata (e.g., [10, 14, 23]) have been applied in the embedded software domain, but also to the schedulability of tasks in concurrent objects [19]. That work complements ours as it does not consider resource restrictions on the concurrency model, but associates deadlines with method calls.
Work on object-oriented models with resource constraints is more scarce. Based on a UML profile for schedulability, performance and time, the informally defined Core Scenario Model (CSM) [28] targets questions in performance model building. CSM has a notion of resource context, which reflects the set of resources used by an operation. CSM aims to bridge the gap between UML specifications and techniques to generate performance models [6]. UML models with stochastic annotations for performance prediction have been proposed for components [16]. Closer to our work is a VDM++ extension to simulate embedded real-time systems [31], in which architectures are explicitly modeled using CPUs and buses, and resources statically bound to the CPUs. However, their work does not address relocation and load balancing strategies.

7 Discussion and Future Work

As software is increasingly developed to be deployed on a variety of architectures, it is important to be able to analyze the behavior of a model under different resource assumptions. ABS uses deployment components with parametric resources to express deployment scenarios for high-level executable models. This paper proposes a primitive for relocating concurrent objects between deployment components, at the abstraction level of the ABS modeling language, which integrates with the formal framework of deployment components in an elegant and simple way. Furthermore, we consider the problem of modeling systems with different load balancing strategies by allowing objects to move between deployment components, depending on the work load of their component. We demonstrate how a simple language extension is sufficient to naturally express dynamic object relocation strategies in this setting; our example shows how traffic on deployment components may cause congestion in the model, resulting in performance degradation for given deployment scenarios, and how load balancing strategies can be used to dynamically alleviate the congestion and thus to improve the overall performance of the model in a given deployment scenario.

For simplicity, this paper uses a quite abstract and simple cost model, ignoring the costs of expression evaluation and communication. We see three different ways in which this cost model may be improved. First, an explicit cost statement could be added to the modeling language, similar to duration statements in timed modeling languages. This leaves the responsibility of correctly capturing the resource consumption for a specific model with the developer. Second, cost profiles could be included to mimic specific architectures. This way, the model could easily be analyzed for different profiles (or cost scenarios), which is difficult to do following the first approach. Third, the cost model could be deduced using static analysis techniques to better approximate the actual cost of communication and evaluation of expressions (e.g., following [3]). Currently, we envisage to follow this approach as it keeps the resource analysis of the models more abstract by overapproximating actual architectures. We have applied this approach for memory analysis of ABS models [4]. Our next step is the integration of static analysis for execution cost with the deployment components with
concurrent resources. Furthermore, a recently developed Java code generator from ABS models opens up the possibility of comparing model-predicted behavior against running code and observing how resource analysis carries over from abstract executable models to the generated code. This is another interesting extension of the modeling framework considered in this paper.

References

The Logic of XACML

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Abstract We study the international standard XACML 3.0 for describing security access control policy in a compositional way. Our main contribution is to derive a logic that precisely captures the idea behind the standard and to formally define the semantics of the policy combining algorithms of XACML. To guard against modelling artifacts we provide an alternative way of characterizing the policy combining algorithms and we formally prove the equivalence of these approaches. This allows us to pinpoint the shortcoming of previous approaches to formalization based either on Belnap logic or on $\mathcal{D}$-algebra.

1 Introduction

XACML (eXtensible Access Control Markup Language) is an approved OASIS\textsuperscript{1} Standard access control language [1,14]. XACML describes both an access control policy language and a request/response language. The policy language is used to express access control policies (who can do what when) while the request/response language expresses queries about whether a particular access should be allowed (requests) and describes answers to those queries (responses).

In order to manage modularity in access control, XACML constructs policies into several components, namely PolicySet, Policy, and Rule. A PolicySet is a collection of others PolicySets or Policies whereas a Policy consists of one or more Rules. A Rule is the smallest component of XACML policy and each Rule only either grants or denies an access. As an illustration, suppose we have access control policies used within the National Health Care System. The system is composed of several access control policies of local hospitals. Each local hospital has its own policies such as patient policy, doctor policy, administration policy, etc. Each policy contains one or more particular rules, for example, in patient policy there is a rule that only the designated patient can read his or her record. In this illustration, both the National Health Care System and local hospital policies are PolicySets. However the patient policy is a Policy and one of its rules is the patient record policy. Every policy is only applicable to a certain target and a policy is applicable when a request matches to its target, otherwise, it is not

\textsuperscript{1} OASIS (Organization for the Advancement of Structured Information Standard) is a non-for-profit, global consortium that drives the development, convergence, and adoption of e-business standards. Information about OASIS can be found at http://www.oasis-open.org.
The evaluation of composing policies is based on a combining algorithm – the procedure for combining decisions from multiple policies. There are four standard combining algorithms in XACML i.e., (i) permit-overrides, (ii) deny-overrides, (iii) first-applicable and (iv) only-one-applicable.

The syntax of XACML is based on XML format [2], while its standard semantics is described normatively using natural language in [14]. Using English paragraphs in standardization leads to misinterpretation and ambiguity. In order to avoid this drawback, we define an abstract syntax of XACML 3.0 and a formal XACML components evaluation based on XACML 3.0 specification in Section 2. Furthermore, the evaluation of the XACML combining algorithms is explained in Section 3.

Recently there are some approaches to formalizing the semantics of XACML. In [8], Halpern and Weissman show XACML formalization using First Order Logic (FOL). However, their formalization does not capture whole XACML specification. It is too expensive to express XACML combining algorithms in FOL. Kolovski et al. in [10,11] maps a large fragment of XACML to Description Logic (DL) – a subset of FOL – but they leave out the formalization of only-one-applicable combining algorithm. Another approach is to represent XACML policies in term of Answer Set Programming (ASP). Although Ahn et al. in [3] show a complete XACML formalization in ASP, their formalization is based on XACML 2.0, which is out-of-date nowadays. More particular, the combining algorithms evaluation in XACML 2.0 is simpler than XACML 3.0. Our XACML 3.0 formalization is closer to multi-valued logic approach such as Belnap logic [4] and $D$-algebra [13]. Bruns et al. in [5,6] and Ni et al. in [13] define a logic for XACML using Belnap logic and $D$-algebra, respectively. In some cases, both works show different results from the XACML standard specification. We discuss the shortcoming of formalization based either on Belnap logic or on $D$-algebra in Section 4 and we conclude in Section 5.

2 XACML Components

XACML syntax is describe verbosely in XML format. For our analysis purpose, we do abstracting XACML components. From the abstraction XACML, we show how XACML evaluates policies. We give an example how XACML policies can be described in our abstraction and the components evaluation at the end of this section.

2.1 Abstracting XACML Components

There are three main policy components in XACML, namely PolicySet, Policy and Rule. PolicySet is the root of all XACML policies. A PolicySet is composed of a sequence of others PolicySet or Policy components along with a policy combining algorithm ID and a Target. A Policy is composed of a sequence of Rule, a Target and a rule combining algorithm ID. A Rule is a single entity that defines the individual rule in the policy. A Rule is composed of a Target, a Condition and its effect, i.e., either deny or permit. A Target is an XACML component that indicates under which categories an XACML policy is applicable. A Target consists of conjunction of AnyOf component with each AnyOf consists of
disjunction of AllOf components and each AllOf consists of conjunction of Match. Each Match contains only one particular category to be matched with the request. Typical categories of XACML attributes are subject category (e.g. human user, workstation, etc) action category (e.g. read, write, delete, etc), resource category (e.g. database, server, etc) and environment category (e.g. SAML, J2SE, CORBA, etc). A Condition is a set of propositional formulae that refines the applicability of a Rule.

A Request contains a set of available informations on desired access request such as subject, action, resource and environment categories. A Request also contains additional information about external state, e.g. the current time, the temperature, etc.

We present in Table 1 a succinct syntax of XACML 3.0 that is faithful to the more verbose syntax used in the standard [14].

<table>
<thead>
<tr>
<th>XACML Policy Components</th>
</tr>
</thead>
<tbody>
<tr>
<td>PolicySet ::= (Target, {PolicySet₁,...,PolicySetₘ}, θ)</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>Policy ::= (Target, {Rule₁,...,Ruleₘ}, θ)</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>Rule ::= (Effect, Target, Condition)</td>
</tr>
<tr>
<td>Condition ::= propositional formulae</td>
</tr>
<tr>
<td>Target ::= Null</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>AnyOf ::= AllOf₁ ∨ ...∧ AllOfₘ</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>AllOf ::= Match₁ ∧ ...∧ Matchₘ</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>Match ::= φ(α)</td>
</tr>
<tr>
<td>φ ::= subject</td>
</tr>
<tr>
<td>α ::= attribute value</td>
</tr>
<tr>
<td>θ ::= p − o</td>
</tr>
<tr>
<td>Effect ::= d</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>XACML Request Component</th>
</tr>
</thead>
<tbody>
<tr>
<td>Request ::= {A₁,...,Aₘ}</td>
</tr>
<tr>
<td></td>
</tr>
<tr>
<td>A ::= φ(α)</td>
</tr>
</tbody>
</table>

### 2.2 XACML Evaluation

The evaluation of XACML components starts from Match evaluation and it is continued iteratively until the PolicySet evaluation. The Match, AllOf, AnyOf, and Target values are either match, not match or indeterminate. The value can be indeterminate if there is an error during the evaluation so that the decision cannot be made at that moment. The Rule evaluation depends on Target evaluation and Condition evaluation. The Condition component is a set of propositional formulae which each formula is evaluated to either true, false or indeterminate. An empty Condition is always evaluated to true. The result of Rule is either applicable, not applicable or indeterminate. An applicable Rule has effect either deny or permit. Finally, the evaluation of Policy and PolicySet are based on a combining algorithm of which the result can be either applicable (with its effect either deny or permit), not applicable or indeterminate.
2.2.1 Three-Valued Lattice

We use three-valued logic to determine XACML evaluation value. We define $L_3 = \langle V_3, \leq \rangle$ be three-valued lattice where $V_3$ is the set \{ $\top$, $I$, $\bot$ \} and $\bot \leq I \leq \top$. Given a subset $S$ of $V_3$, we denote the greatest lower bound (glb) and the least upper bound (lub) at $S$ (w.r.t. $L_3$) by $\bigcap S$ and $\bigcup S$, respectively. Recall that $\bigcap \emptyset = \top$ and $\bigcup \emptyset = \bot$.

We use $\llbracket.$ notation to map XACML elements into their evaluation values. The evaluation of XACML components to values in $V_3$ is summarized in Table 2.

Table 2. Mapping $V_3$ into XACML Evaluation Values

<table>
<thead>
<tr>
<th>$V_3$</th>
<th>Match and Target value</th>
<th>Condition value</th>
<th>Rule, Policy and PolicySet value</th>
</tr>
</thead>
<tbody>
<tr>
<td>$\top$</td>
<td>match</td>
<td>true</td>
<td>applicable (either deny or permit)</td>
</tr>
<tr>
<td>$\bot$</td>
<td>not match</td>
<td>false</td>
<td>not applicable</td>
</tr>
<tr>
<td>$I$</td>
<td>indeterminate</td>
<td>indeterminate</td>
<td>indeterminate</td>
</tr>
</tbody>
</table>

2.2.2 Match Evaluation

A Match element $M$ is an attribute value that the request should fulfill. Given a Request component $Q$, the evaluation of Match element is as follows:

$$\llbracket M \rrbracket(Q) = \begin{cases} \top & M \in Q \\ \bot & M \notin Q \\ I & \text{there is an error during the evaluation} \end{cases} \quad (1)$$

2.2.3 Target Evaluation

Let $M$ be a Match, $A = M_1 \land \ldots \land M_m$ be an AllOf, $E = A_1 \lor \ldots \lor A_n$ be an AnyOf, $T = E_1 \land \ldots \land E_o$ be a Target and $Q$ be a Request. Then, the evaluations of AllOf, AnyOf, and Target are as follows:

$$\llbracket A \rrbracket(Q) = \bigcap_{i=1}^{m} \llbracket M_i \rrbracket(Q) \quad (2)$$

$$\llbracket E \rrbracket(Q) = \bigcup_{i=1}^{n} \llbracket A_i \rrbracket(Q) \quad (3)$$

$$\llbracket T \rrbracket(Q) = \bigcap_{i=1}^{o} \llbracket E_i \rrbracket(Q) \quad (4)$$

In summary, we can simplify the Target evaluation as follows:

$$\llbracket T \rrbracket(Q) = \bigcap \bigcup \bigcap \llbracket M \rrbracket(Q) \quad (5)$$

An empty Target – indicated by $\text{Null}$ – is always evaluated to $\top$. 

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2.2.4 Condition Evaluation

We define the conditional evaluation function \( eval \) as an arbitrary function to evaluate \( Condition \) to value in \( V_3 \) given a \( Request \) component \( Q \). The evaluation of \( Condition \) is defined as follows:

\[
[C](Q) = eval(C, Q) \quad (6)
\]

2.2.5 Extended Values

In order to distinguish an applicable policy to permit an access from applicable policy to deny an access, we extend \( \top \) in \( V_3 \) value to \( \top_p \) and \( \top_d \), respectively. The same case also applies to indeterminate value. The extended indeterminate value contains the potential effect values which could have occurred if there would not have been an error during a evaluation. The possible extended indeterminate values are [14]:

- Indeterminate Deny (\( I_d \)): an indeterminate from a policy which could have evaluated to deny but not permit, e.g., a Rule which evaluates to indeterminate and its effect is deny.
- Indeterminate Permit (\( I_p \)): an indeterminate from a policy which could have evaluated to permit but not deny, e.g., a Rule which evaluates to indeterminate and its effect is permit.
- Indeterminate Deny Permit (\( I_{dp} \)): an indeterminate from a policy which could have effect either deny or permit.

We extend the set \( V_3 \) to \( V_6 = \{ \top_p, \top_d, I_d, I_p, I_{dp}, \bot \} \) and we use \( V_6 \) for for XACML policies evaluations.

2.2.6 Rule Evaluation

Let \( R = \langle *, T, C \rangle \) be a Rule and \( Q \) be a Request. Then, the evaluation of \( Rule \) is determined as follows:

\[
[R](Q) = \begin{cases} 
\top & \text{[}\top](Q) = \top \text{ and } [C](Q) = \top \\
\bot & \text{[}\top](Q) = \top \text{ and } [C](Q) = \bot \text{ or } [\bot](Q) = \bot \\
I & \text{otherwise}
\end{cases} \quad (7)
\]

Let \( F \) and \( G \) be two values in \( V_3 \). We define a new operator \( \sim \): \( V_3 \times V_3 \to V_3 \) as follows:

\[
F \sim G = \begin{cases} 
G & F = \top \\
F & \text{otherwise}
\end{cases} \quad (8)
\]

We define a function \( \sigma : V_3 \times \{ p, d \} \to V_6 \) that maps a value in \( V_3 \) into a value in \( V_6 \) given a particular \( Rule \)'s effect as follows:

\[
\sigma(X, *) = \begin{cases} 
X & X = \bot \\
X & \text{otherwise}
\end{cases} \quad (9)
\]

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Proposition 1. Let $\mathcal{R} = \langle *, T, C \rangle$ be a Rule and $Q$ be a Request. Then, the following equation holds

$$\llbracket \mathcal{R} \rrbracket(Q) = \sigma(\llbracket T \rrbracket(Q) \mapsto \llbracket C \rrbracket(Q), *)$$

(10)

2.2.7 Policy Evaluation

The standard evaluation of Policy element taken from [14] is as follows

<table>
<thead>
<tr>
<th>Target value</th>
<th>Rule value</th>
<th>Policy Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>match</td>
<td>At least one Rule value is applicable</td>
<td>Specified by the combining algorithm</td>
</tr>
<tr>
<td>match</td>
<td>All Rule values are not applicable</td>
<td>not applicable</td>
</tr>
<tr>
<td>not match</td>
<td>At least one Rule value is indeterminate</td>
<td>Specified by the combining algorithm</td>
</tr>
<tr>
<td>indeterminate</td>
<td>Don’t care</td>
<td>not applicable</td>
</tr>
</tbody>
</table>

Let $P = \langle T, R, \theta \rangle$ be a Policy where $R = \langle R_1, \ldots, R_n \rangle$, Let $Q$ be a Request and $R' = (\llbracket R_1 \rrbracket(Q), \ldots, \llbracket R_n \rrbracket(Q))$. The evaluation of Policy is defined as follows:

$$\llbracket P \rrbracket(Q) = \begin{cases} I* & \llbracket T \rrbracket(Q) = I \text{ and } \bigoplus_\theta(R') \in \{T*, I*\} \\ \bot & \llbracket T \rrbracket(Q) = \bot \text{ or } \llbracket T \rrbracket(Q) = T \text{ and } \forall R_i : \llbracket R_i \rrbracket(Q) = \bot \\ \bigoplus_\theta(R') & \text{otherwise} \end{cases}$$

(11)

Note 1. The combining algorithms denoted by $\bigoplus$ is explained in Section 3.

2.2.8 PolicySet Evaluation

The evaluation of PolicySet is similar to Policy evaluation. However, the input of the combining algorithm is a sequence of either PolicySet or Policy components.

Let $PS = \langle T, P, \theta \rangle$ be a PolicySet where $P = \langle P_1, \ldots, P_n \rangle$. Let $Q$ be a Request and $P' = (\llbracket P_1 \rrbracket(Q), \ldots, \llbracket P_n \rrbracket(Q))$. The evaluation of PolicySet is defined as follows:

$$\llbracket PS \rrbracket(Q) = \begin{cases} I* & \llbracket T \rrbracket(Q) = I \text{ and } \bigoplus_\theta(P') \in \{T*, I*\} \\ \bot & \llbracket T \rrbracket(Q) = \bot \text{ or } \llbracket T \rrbracket(Q) = T \text{ and } \forall P_i : \llbracket P_i \rrbracket(Q) = \bot \\ \bigoplus_\theta(P') & \text{otherwise} \end{cases}$$

(12)

2.3 Examples

The following examples simulate briefly how a policy is built using the abstraction. The examples are motivated by [7,9] which presents a health information system for a small nursing home in New South Wales, Australia.

Example 1 (Patient Policy). The general policy in the hospital in particular:
1. Patient Record Policy
   - RP1: only designated patient can read his or her patient record except that if
     the patient is less than 18 years old, the patient’s guardian is permitted also
     read the patient’s record,
   - RP2: patients may only write patient surveys into their own records
   - RP3: both doctors and nurses are permitted to read any patient records,

2. Medical Record Policy
   - RM1: doctors may only write medical records for their own patients and
   - RM2: may not write any other patient records,

The XACML policies for this example is shown in Figure 1. The topmost policy in
this example is the Patient Policy that contains two policies, namely the Patient
Record Policy and the Medical Record Policy. The access is granted if either one of the Patient
Record Policy or the Medical Record Policy gives a permit access. Thus in this case,
we use permit-overrides combining algorithm to combine those two policies. In order
to restrict the access, each policy denies an access if there is a rule denies it. Thus, we
use deny-overrides combining algorithms to combine the rules.

```
PS_patient = <Null, <P_patient_record, P_medical_record>, po>
P_patient_record = <Null, <RP1, RP2, RP3>, do>
P_medical_record = <Null, <RM1, RM2>, do>

RP1 = < p, subject(patient) \ action(read) \ resource(patient_record),
   patient(id,X) \ patient_record(id,Y) \ (X = Y | (age(Y) < 18 | guardian(X,Y))>

RP2 = < p, subject(patient) \ action(write) \ resource(patient_survey),
   patient(id,X) \ patient_survey(id, X)>

RP3 = < p, subject(doctor) \ subject(nurse) \ action(read) \ resource(patient_record),
   true>

RM1 = < d, subject(doctor) \ action(write) \ resource(medical_record),
   doctor(id,X) \ patient(id,Y) \ medical_record(id, Y) \ patient_doctor(Y,X)>

RM1 = < d, subject(doctor) \ action(write) \ resource(medical_record),
   doctor(id,X), patient(id,Y), medical_record(id, Y), not patient_doctor(Y,X)>
```

Figure 1. The XACML Policy for Patient Policy

Suppose now there is an emergency situation and a doctor D asks permission to
read patient record P. The Request is as follows:

```
{ subject(doctor), action(read), resource(patient_record),
   doctor(id,d), patient(id,p), patient_record(id,p))
```
Only Target RP3 matches for this request and the effect of RP3 is permit. Thus, the final result is doctor D is allowed to read patient record P. Now, suppose that after doing some treatment, the doctor wants to update the medical record. A request is sent

{ subject(doctor), action(write), resource(medical_record),
  doctor(id,d), patient(id,p), medical_record(id,p) }

The Target RM1 and the Target RM2 match for this request, however because doctor D is not registered as patient P’s doctor thus Condition RM1 is evaluated to false while Condition RM2 is evaluated to true. In consequence, Rule RM1 is not applicable while Rule RM2 is applicable with effect deny.

3 Combining Algorithms

Currently, there are four basic combining algorithms in XACML, namely (i) permit-overrides, (ii) deny-overrides, (iii) first-applicable, and (iv) only-one-applicable. The input of a combining algorithm is a sequence of Rule, Policy or PolicySet values. In this section we give formalizations of the XACML 3.0 combining algorithms based on [14]. To guard against modelling artifacts we provide an alternative way of characterizing the policy combining algorithms and we formally prove the equivalence of these approaches.2

3.1 Pairwise Policy Values

In $V_6$ we define the truth values of XACML components by extending $\top$ to $\top_p$ and $\top_d$ and $I$ to $I_d$, $I_p$ and $I_{dp}$. This approach shows straightforwardly the status of XACML component. However, it is easier if we use numerical encoding when we need to do a computation, especially for computing policies compositions. Thus, we encode all the values returned by algorithms as pairs of natural numbers.

In this numerical encoding, the value 1 represents an applicable value (either deny or permit), $\frac{1}{2}$ represents indeterminate value and 0 means there is no applicable value. In each tuple, the first element represents the Deny value ($\top_d$) and the later represents Permit value ($\top_p$). We can say $[0, 0]$ for not applicable (⊥) because neither Deny nor Permit is applicable, $[1, 0]$ for applicable with deny effect ($\top_d$) because only Deny value is applicable, $[\frac{1}{2}, 0]$ for $I_d$ because the Deny part is indeterminate, $[\frac{1}{2}, \frac{1}{2}]$ for $I_{dp}$ because both Deny and Permit have indeterminate values. The conversion applies also for Permit.

A set of pairwise policy values is $P = \{ [0, 0], [\frac{1}{2}, 0], [0, \frac{1}{2}], [1, 0], [\frac{1}{2}, \frac{1}{2}], [0, 1] \}$. Let $[D, P]$ be an element on $P$. We denote $d([D, P]) = D$ and $p([D, P]) = P$ for the function that returns the Deny value and Permit value, respectively.

An extended version of this paper with all the proofs is available at http://www2.imm.dtu.dk/~cdpu/Papers/the_logic_of_XACML-extended.pdf.
We define $\delta: V_6 \rightarrow P$ as a mapping function that maps $V_6$ into $P$ as follows:

$$
\delta(X) = \begin{cases}
[0, 0] & X = \perp \\
[1, 0] & X = \top_d \\
[0, 1] & X = \top_p \\
[\frac{1}{2}, 0] & X = I_d \\
[0, \frac{1}{2}] & X = I_p \\
[\frac{1}{2}, \frac{1}{2}] & X = I_{dp}
\end{cases}
$$

We define $\delta$ over a sequence $S$ as $\delta(S) = \langle \delta(s) \mid s \in S \rangle$.

We use pairwise comparison for the order of $P$. We define an order $\sqsubseteq_P$ for $P$ as follows $[D_1, P_1] \sqsubseteq_P [D_2, P_2]$ iff $D_1 \leq D_2$ and $P_1 \leq P_2$ with $0 \leq \frac{1}{2} \leq 1$. We write $P_P$ for the partial ordered set (poset) $(P, \sqsubseteq_P)$ illustrated in Figure 2.

![Figure 2. The Partial Ordered Set $P_P$ for Pairwise Policy Values](image)

Let $\max: 2^{\mathbb{R}} \rightarrow \mathbb{R}$ be a function that returns the maximum value of a set of rational numbers and let $\min: 2^{\mathbb{R}} \rightarrow \mathbb{R}$ be a function that returns the minimum value of a set of rational numbers. We define $\operatorname{Max}_{\sqsubseteq_P}: 2^P \rightarrow P$ as a function that returns the maximum pairwise policy value which is defined as follows:

$$
\operatorname{Max}_{\sqsubseteq_P}(S) = [\max(\{d(X) \mid X \in S\}), \max(\{p(X) \mid X \in S\})]
$$

and $\operatorname{Min}_{\sqsubseteq_P}: 2^P \rightarrow P$ as a function that return the minimum pairwise policy value which is defined as follows:

$$
\operatorname{Min}_{\sqsubseteq_P}(S) = [\min(\{d(X) \mid X \in S\}), \min(\{p(X) \mid X \in S\})]
$$

### 3.2 Permit-Overrides Combining Algorithm

The permit-overrides combining algorithm is intended for those cases where a permit decision should have priority over a deny decision. This algorithm (taken from [14]) has the following behaviour:

1. If any decision is $\top_p$ then the result is $\top_p$. 

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Figure 3. The Lattice $L_{p-o}$ for The Permit-Overrides Combining Algorithm (left), The Lattice $L_{d-o}$ for The Deny-Overrides Combining Algorithm (middle) and The Lattice $L_{o-1-o}$ for The Only-One-Applicable Combining Algorithm (right)

2. otherwise, if any decision is $I_{dp}$ then the result is $I_{dp}$,
3. otherwise, if any decision is $I_p$ and another decision is $I_d$ or $\top_d$, then the result is $I_{dp}$,
4. otherwise, if any decision is $I_d$ then the result is $I_d$,
5. otherwise, if decision is $\top_d$ then the result is $\top_d$,
6. otherwise, if any decision is $I_d$ then the result is $I_d$,
7. otherwise, the result is $\bot$.

We call $L_{p-o} = (V_6, \sqsubseteq_{p-o})$ for the lattice using the permitoverrides combining algorithm where $\sqsubseteq_{p-o}$ is the ordering depicted in Figure 3. The least upper bound operator for $L_{p-o}$ is denoted by $\bigcup_{p-o}$.

Definition 1. The permit-overrides combining algorithm $\biguplus_{V_6}^{p-o}$ is a mapping function from a sequence of $V_6$ elements into an element in $V_6$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $V_6$ and $S' = \{ s_1, \ldots, s_n \}$. We define the permit-overrides combining algorithm under $V_6$ as follows:

$$\biguplus_{V_6}^{p-o}(S) = \bigcup_{p-o} S'$$

(16)

The permit-overrides combining algorithm can also be expressed under $P$. The idea is that we inspect the maximum value of Deny and Permit in the set of pairwise policy values. We conclude that the decision is permit if the Permit is applicable (i.e. it has value 1). If the Permit is indeterminate (i.e. it has value $\frac{1}{2}$) then the decision is $I_{dp}$ if the Deny is either indeterminate (i.e. it has value $\frac{1}{2}$) or applicable (i.e. it has value 1). Otherwise we take the maximum value of Deny and Permit from the set of pairwise policy values as the result of permit-overrides combining algorithm.

Definition 2. The permit-overrides combining algorithm $\biguplus_{P-0}^{p-o}$ is a mapping function from a sequence of $P$ elements into an element in $P$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of pairwise policy values and $S' = \{ s_1, \ldots, s_n \}$. 

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We define the permit-overrides combining algorithm under $\mathbb{P}$ as follows:

\[
\bigoplus_{\mathbb{P}}(S) = \begin{cases} 
[0, 1] & \text{Max}_{\mathbb{P}}(S') = [\cdot, 1] \\
[\frac{1}{2}, \frac{3}{2}] & \text{Max}_{\mathbb{P}}(S') = [D, \frac{1}{2}], D \geq \frac{1}{2} \\
\text{Max}_{\mathbb{P}}(S') & \text{otherwise}
\end{cases}
\]  

(17)

Proposition 2. Let $S$ be a sequence of policy values in $V_6$. Then

\[
d_{\mathbb{P}}(\bigoplus_{\mathbb{P}}(S)) = \bigoplus_{\mathbb{P}}(\delta(S))
\]

3.3 Deny-Overrides Combining Algorithm

The deny-overrides combining algorithm is intended for those cases where a deny decision should have priority over a permit decision. This algorithm (taken from [14]) has the following behaviour:

1. If any decision is $\top_d$ then the result is $\top_d$,
2. otherwise, if any decision is $I_{dp}$ then the result is $I_{dp}$,
3. otherwise, if any decision is $I_d$ and another decision is $I_p$ or $\top_p$, then the result is $I_{dp}$,
4. otherwise, if any decision is $I_d$ then the result is $I_d$,
5. otherwise, if decision is $\top_p$ then the result is $\top_p$,
6. otherwise, if any decision is $I_p$ then the result is $I_p$,
7. otherwise, the result is $\bot$.

We call $L_{d-o} = (V_6, \sqsubseteq_{d-o})$ for the lattice using the deny-overrides combining algorithm where $\sqsubseteq_{d-o}$ is the ordering depicted in Figure 3. The least upper bound operator for $L_{d-o}$ is denoted by $\bigoplus_{d-o}$.

Definition 3. The deny-overrides combining algorithm $\bigoplus_{d-o}^{V_6}$ is a mapping function from a sequence of $V_6$ elements into an element in $V_6$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $V_6$ and $S' = \{ s_1, \ldots, s_n \}$. We define the deny-overrides combining algorithm under $V_6$ as follows:

\[
\bigoplus_{d-o}^{V_6}(S) = \bigsqcup_{d-o} S'
\]  

(18)

The deny-overrides combining algorithm can also be expressed under $\mathbb{P}$. The idea is similar to permit-overrides combining algorithm by symmetry.

Definition 4. The deny-overrides combining algorithm $\bigoplus_{d-o}^{\mathbb{P}}$ is a mapping function from a sequence of $\mathbb{P}$ elements into an element in $\mathbb{P}$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $\mathbb{P}$ and $S' = \{ s_1, \ldots, s_n \}$. We define the deny-overrides combining algorithm under $\mathbb{P}$ as follows:

\[
\bigoplus_{d-o}^{\mathbb{P}}(S) = \begin{cases} 
[1, 0] & \text{Max}_{\mathbb{P}}(S') = [\cdot, 1] \\
[\frac{1}{2}, \frac{3}{2}] & \text{Max}_{\mathbb{P}}(S') = [\frac{1}{2}, P], P \geq \frac{1}{2} \\
\text{Max}_{\mathbb{P}}(S') & \text{otherwise}
\end{cases}
\]  

(19)
Proposition 3. Let $S$ be a sequence of policy values in $V_6$. Then

$$\delta(\bigoplus_{d-o}(S)) = \bigoplus_{d-o}(\delta(S))$$

3.4 First-Applicable Combining Algorithm

The result of first-applicable algorithm is the first Rule, Policy or PolicySet element in the sequence whose Target and Condition is applicable. The pseudo-code of the first-applicable combining algorithm in XACML 3.0 [14] shows that the result of this algorithm is the first Rule, Policy or PolicySet that is not "not applicable". The idea is that there is a possibility an indeterminate policy could return to be an applicable policy. The first-applicable combining algorithm under $V_6$ and $P$ are defined below.

Definition 5 (First-Applicable Combining Algorithm). The first-applicable combining algorithm $\bigoplus_{V_6}$ is a mapping function from a sequence of $V_6$ elements into an element in $V_6$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $V_6$. We define the first-applicable combining algorithm under $V_6$ as follows:

$$\bigoplus_{V_6}(S) = \begin{cases} s_i & \exists i : s_i \neq \bot \text{ and } \forall j < i : s_j = \bot \\ \bot & \text{otherwise} \end{cases} \quad (20)$$

Definition 6. The first-applicable combining algorithm $\bigoplus_{P}$ is a mapping function from a sequence of $P$ elements into an element in $P$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $P$. We define the first applicable combining algorithm under $P$ as follows:

$$\bigoplus_{P}(S) = \begin{cases} s_i & \exists i : s_i \neq [0,0] \text{ and } \forall j < i : s_j = [0,0] \\ [0,0] & \text{otherwise} \end{cases} \quad (21)$$

Proposition 4. Let $S$ be a sequence of policy values in $V_6$. Then

$$\delta(\bigoplus_{d-o}(S)) = \bigoplus_{d-o}(\delta(S))$$

3.5 Only-One-Applicable Combining Algorithm

The result of the only-one-applicable combining algorithm ensures that one and only one policy is applicable by virtue of their Target. If no policy applies, then the result is not applicable, but if more than one policy is applicable, then the result is indeterminate. When exactly one policy is applicable, the result of the combining algorithm is the result of evaluating the single applicable policy.

We call $L_{o-1-a} = (V_6, \sqsubseteq_{o-1-a})$ for the lattice using the only-one-applicable combining algorithm where $\sqsubseteq_{o-1-a}$ is the ordering depicted in Figure 3. The least upper bound operator for $L_{o-1-a}$ is denoted by $\bigcup_{o-1-a}$. 

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Definition 7. The only-one-applicable combining algorithm $\bigoplus_{a=1-a}^{V_6}$ is a mapping function from a sequence of $V_6$ elements into an element in $V_6$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $V_6$ and $S' = \{ s_1, \ldots, s_n \}$. We define only-one-applicable combining algorithm under $V_6$ as follows:

$$\bigoplus_{a=1-a}^{V_6}(S) = \begin{cases} I_d & \exists i, j : i \neq j, s_i = s_j = \top_d \text{ and } \\
I_p & \forall k, s_k \neq \top_d \rightarrow s_k = \bot \\
\bigcup_{a=1-a}^{S'} & \exists i, j : i \neq j, s_i = s_j = \top_p \text{ and } \\
& \forall k, s_k \neq \top_p \rightarrow s_k = \bot \\
\end{cases}$$

The only-one-applicable combining algorithm also can be expressed under $P$. The idea is that we inspect the maximum value of Deny and Permit returned from the given set of pairwise policy values. By inspecting the maximum value for each element, we know exactly the combination of pairwise policy values i.e., if we find that both Deny and Permit are not 0, it means that the Deny and the Permit are either applicable (i.e. it has value 1) or indeterminate (i.e. it has value $\frac{1}{2}$). Thus, the result of this algorithm is $I_{dp}$ (based on the XACML 3.0 Specification [14]). However if only one element is not 0 then there is a possibility that many policies have the same applicable (or indeterminate) values. If there are at least two policies with the Deny (or Permit) are either applicable or indeterminate value, then the result is $I_d$ (or $I_p$). Otherwise we take the maximum value of Deny and Permit from the given set of pairwise policy values as the result of only-one-applicable combining algorithm.

Definition 8. The only-one-applicable combining algorithm $\bigoplus_{a=1-a}^{P}$ is a mapping function from a sequence of $P$ elements into an element in $P$ as the result of composing policies. Let $S = \langle s_1, \ldots, s_n \rangle$ be a sequence of policy values in $P$ and $S' = \{ s_1, \ldots, s_n \}$. We define only-one-applicable combining algorithm under $P$ as follows:

$$\bigoplus_{a=1-a}^{P}(S) = \begin{cases} [\frac{1}{2}, \frac{1}{2}] & \max_{\leq P}(S') = [D, P], D, P \geq \frac{1}{2} \\
[\frac{1}{2}, 0] & \max_{\leq P}(S') = [D, 0], D \geq \frac{1}{2} \text{ and } \\
[0, \frac{1}{2}] & \exists i, j : i \neq j, d(s_i), d(s_j) \geq \frac{1}{2} \text{ and } \\
\max_{< P}(S') & \exists i, j : i \neq j, p(s_i), p(s_j) \geq \frac{1}{2} \\
& \max_{\leq P}(S') \text{ otherwise} \\
\end{cases}$$

Proposition 5. Let $S$ be a sequence of policy values in $V_6$. Then

$$\delta(\bigoplus_{a=1-a}^{V_6}(S)) = \bigoplus_{a=1-a}^{P}(\delta(S))$$

4 Related Work

We will focus the discussion on the formalization of XACML using Belnap logic [4] and D-Algebra [13] – those two have a similar approach to the pairwise policy values.
approach explained in Section 3. We show the shortcoming of the formalization on Bruns et al. work in [6] and Ni et al. work in [13].

4.1 XACML Semantics under Belnap Four-Valued Logic

Belnap in his paper [4] defines a four-valued logic over four = \{\top\top, \top\bot, \bot\top, \bot\bot\}. There are two orderings in Belnap logic, i.e., the knowledge ordering (≤k) and the truth ordering (≤t) (see Figure 4).

![Figure 4. Bi-lattice of Belnap Four-Valued Logic](image)

Bruns et al. in PBel [5,6] and also Hankin et al. in AspectKB [9] use Belnap four-valued logic to represent the composition of access control policies. The responses of an access control system are

- permit-overrides: \((p \oplus B q)[\top\top \mapsto \bot\bot]\)
- first-applicable: \(p > q\)
- only-one-applicable: \(p \oplus B q \oplus B (p \oplus B \neg p) \oplus B (q \oplus B \neg q)\)

Bruns et al. defined XACML combining algorithms using Belnap four-valued logic as follows [6]:

- overwriting operator \([y \mapsto z]\) with \(y, z \in \text{four}\). Expression \(x[y \mapsto z]\) yields \(x\) if \(x \neq y\), and \(z\) otherwise.
- priority operator \(x > y\); it is a syntactic sugar of \(x[\bot\bot \mapsto y]\).

Bruns et al. suggested that the indeterminate value is treated as \(\top\top\). However, with indeterminate as \(\top\top\), the permit-overrides combining algorithm is not defined correctly. Suppose we have two policies: \(p\) and \(q\) where \(p\) is permit and \(q\) is indeterminate. The result of the permit-overrides combining algorithm is as follows \((p \oplus B q)[\top\top \mapsto \bot\bot]\) = \((\top \oplus B \top\top)[\top\top \mapsto \bot\bot] = \top\top[\top\top \mapsto \bot\bot] = \bot\bot\). Based on the XACML 2.0 [12] and the XACML 3.0 [14], the result of permit-overrides combining algorithm should be permit (\(\top\top\)). However, based on Belnap four-valued logic, the result is deny (\(\bot\bot\)).

Bruns et al. tried to define indeterminate value as a conflict by formalizing it as \(\top\top\). However, their formulation of permit-overrides combining algorithm is inconsistent based on the standard XACML specification. Moreover, they said that sometimes
indeterminate should be treated as ⊥ and sometimes as ⊤ [5], but there is no explanation about under which circumstances that indeterminate is treated as ⊤ or as ⊥. The treatment of indeterminate as ⊤ is too strong because indeterminate does not always contain information about deny and permit in the same time. Only \( I_d \) contains information both deny and permit. However, \( I_d \) and \( I_p \) only contain information only about deny and permit, respectively. Even so, the value ⊥ for indeterminate is too weak because indeterminate is treated as not applicable despite that there is information contained inside indeterminate value. The Belnap four-valued logic has no explicit definition of indeterminate. In contrast, the Belnap four-valued has a conflict value (i.e. ⊤).

4.2 XACML Semantics under \( \mathcal{D} \)-Algebra

Ni et al. in [13] define \( \mathcal{D} \)-algebra as a decision set together with some operations on it.

**Definition 9 (\( \mathcal{D} \)-algebra [13]).** Let \( D \) be a nonempty set of elements, 0 be a constant element of \( D \), \( ¬ \) be a unary operation on elements in \( D \), and \( ⊕, ⊗ \) be binary operations on elements in \( D \). A \( \mathcal{D} \)-algebra is an algebraic structure \( \langle D, ¬, ⊕, ⊗, 0 \rangle \) closed on \( ¬, ⊕, ⊗ \) and satisfying the following axioms:

1. \( x ⊕ y = y ⊕ x \)
2. \( (x ⊕ y) ⊕ z = x ⊕ (y ⊕ z) \)
3. \( x ⊕ 0 = x \)
4. \( ¬¬x = x \)
5. \( x ⊕ ¬0 = ¬0 \)
6. \( ¬(¬x ⊕ y) ⊕ y = ¬(¬y ⊕ x) ⊕ x \)
7. \( x ⊗ y = \begin{cases} ¬0 & : x = y \\ 0 & : x ≠ y \end{cases} \)

In order to write formulae in a compact form, for \( x, y ∈ D \), \( x ⊗ D y = ¬(¬x ⊕ D ¬y) \) and \( x ⊕ D y = x ⊕ D ¬y \).

Ni et al. [13] show that XACML decisions contain three different value, i.e. permit (\{p\}), deny (\{d\}) and not applicable (\{n\}). Those decision are deterministic decisions. The non-deterministic decisions such as \( I_d \), \( I_p \) and \( I_{dp} \) are denoted by \{d, n\}, \{p, n\}, and \{d, p, n\}, respectively. The interpretation of a \( \mathcal{D} \)-algebra on XACML decisions is as follows [13]:

- \( D \) is represented by \( \mathcal{P}(\{p, d, n\}) \)
- \( 0 \) is represented by \( \emptyset \)
- \( ¬x \) is represented by \{p, d, n\} \( x \) where \( x ∈ D \)
- \( x ⊕ D y \) is represented by \( x ∪ y \) where \( x, y ∈ D \)
- \( ⊗D \) is defined by axiom 7

There are two values which are not in XACML, i.e. \( \emptyset \) and \{p, d\}. Simply we say \( \emptyset \) for empty policy (or there is no policy) and \{p, d\} for a conflict.
The composition function of permit-overrides using $D$-Algebra is as follows:

$$f_{po}(x, y) = (x \oplus_D y)$$

$$\ominus_D (((x \otimes_D \{ p \}) \oplus_D (y \otimes_D \{ p \})) \oplus_D \{ d, \frac{n}{a} \})$$

$$\ominus_D ((x \otimes_D y) \otimes_D \{ \frac{n}{a} \}) \ominus_D \{ \frac{n}{a} \} \ominus_D -((x \otimes_D \emptyset) \oplus_D (y \otimes_D \emptyset))$$

The composition function that Ni et al. proposed is inconsistent with neither the XACML 3.0 [14] nor the XACML 2.0 [12] as they claimed in [13]. Below we show an example that compares all of the results of permit-overrides combining algorithm under the logics discussed in this paper.

Example 2. Given two policies $P_1$ and $P_2$ where $P_1$ is Indeterminate Permit and $P_2$ is Deny. Let us use the permit-overrides combining algorithm to compose those two policies. Table 3 shows the result of combining polices under Belnap logic, $D$-algebra, $V_6$ and $P$.

<table>
<thead>
<tr>
<th>Logic</th>
<th>$P_1$</th>
<th>$P_2$</th>
<th>Permit-Overrides Function</th>
<th>Result</th>
</tr>
</thead>
<tbody>
<tr>
<td>Belnap logic $D$-algebra</td>
<td>$\top$</td>
<td>$\top$</td>
<td>$\ominus_D { p, \frac{n}{a} }$</td>
<td>$\ominus_D { p, \frac{n}{a} }$</td>
</tr>
<tr>
<td>$V_6$</td>
<td>$\top$</td>
<td>$\top$</td>
<td>$\ominus_D { p, \frac{n}{a} }$</td>
<td>$\ominus_D { p, \frac{n}{a} }$</td>
</tr>
<tr>
<td>$P$</td>
<td>$[0, \frac{1}{2}]$</td>
<td>$[1, 0]$</td>
<td>$\ominus_D { \frac{1}{2}, \frac{1}{2} }$</td>
<td>$\ominus_D { \frac{1}{2}, \frac{1}{2} }$</td>
</tr>
</tbody>
</table>

The result of permit-overrides combining algorithm under Belnap logic is $\top\top$ and under $D$-algebra is $p, \frac{n}{a}$. Under Bruns et al. approach using Belnap logic, the access is denied while under Ni et al. approach using $D$-algebra, a conflict occurs. Both Bruns et al. and Ni et al. claim that their approaches fit with XACML 2.0 [12]. Moreover $D$-algebra claims that it fits with XACML 3.0 [14]. However based on XACML 2.0 the result should be Indeterminate and based on XACML 3.0 the result should be Indeterminate Deny Permit and neither Belnap logic nor $D$-algebra fits the specifications. We have illustrated that Belnap logic and $D$-algebra in some cases give different result with the XACML specification. Conversely, our approach gives consistent result based on the XACML 3.0 [14] and on the XACML 2.0 [12].

5 Conclusion

We have shown the formalization of XACML version 3.0 step by step. We believe that with our approach, the user can understand better about how XACML works especially in the behaviour of combining algorithms. We show two approaches to formalizing standard XACML combining algorithms, i.e., using $V_6$ and $P$. To guard against modeling artifacts, we formally prove the equivalence of these approaches.

The pairwise policy values approach is useful in defining new combining algorithms. For example, suppose we have a new combining algorithm "all permit", i.e., the result...
of composing policies is permit if all policies give permit values, otherwise it is deny. Using pairwise policy values approach the result of composing a set of policies values S is permit ([0,1]) if \( \min_{D} (S) = [0, 1] = \max_{D} (S) \), otherwise, it is deny ([1,0]).

Ni et al. proposes a \( D \)-algebra over a set of decisions for XACML combining algorithms in [13]. However, there are some mismatches between their results and the XACML specifications. Their formulations are inconsistent based both on the XACML 2.0 [12] and on the XACML 3.0 [14].

Both Belnap four-valued logic and \( D \)-Algebra have a conflict value. In XACML, the conflict will never occur because the combining algorithms do not allow that. Conflict value might be a good indication that the policies are not well design. We propose an extended \( P \) which captures a conflict value in Appendix A.

References


\[^{3}\]The detail of all of XACML decisions under \( D \)-algebra can be seen in extended paper at http://www2.imm.dtu.dk/~cdpu/Papers/the_logic_of_XACML-extended.pdf.
A Extended Pairwise Policy Values

We add three values into $P$, i.e. deny with indeterminate permit ([1, 1]), permit with indeterminate deny ([1, 1]) and conflict ([1, 1]) and we call the extended pairwise policy values $P_9 = P \cup \{[1, \frac{1}{2}], [\frac{1}{2}, 1], [1, 1]\}$. The extended pairwise policy values shows all possible combination of pairwise policy values. The ordering of $P_9$ is illustrated in Figure 5.

![Figure 5. Nine-Valued Lattice](image)

We can see that $P_9$ forms a lattice (we call this $L_9$) where the top element is [1, 1] and the bottom element is [0, 0]. The ordering of this lattice is the same as $\sqcap_P$ where the greatest lower bound and the least upper bound for $S \subseteq P_9$ are defined as follows:

$$\bigcap_{L_9} S = Max_{\sqsubseteq_P}(S) \text{ and } \bigcup_{L_9} S = Min_{\sqsubseteq_P}(S)$$
A proof assistant based formalization of
components in MDE

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Abstract. Model driven engineering (MDE) now plays a key role in the development of safety critical systems through the use of early validation and verification of models, and the automatic generation of software and hardware artifacts from the validated and verified models. In order to ease the integration of formal specification and verification technologies, various formalizations of the MDE technologies were proposed by different authors using term or graph rewriting, proof assistants, logical frameworks, etc.

The use of components is also mandatory to improve the efficiency of system development. Invasive Software Composition (ISC) has been proposed by Aßman in [1] to add a generic component structure to existing Domain Specific Modeling Languages in MDE. This approach is the basis of the ReuseWare toolset.

We present in this paper an extension of a formal embedding of some key aspects of MDE in set theory in order to formalize ISC and prove the correctness of the proposed approach with respect to the conformance relation with the base metamodel. The formal embedding we rely on was developed by some of the authors, presented in [23] and then implemented using the Calculus of Inductive Construction and the Coq proof-assistant. This work³ is a first step in the formalization of composable verification technologies in order to ease its integration for DSML extended with component features using ISC.

1 Introduction

Model driven engineering now plays a key role in the development of safety critical systems through the use of model early validation and verification, and the automatic generation of software or hardware artefacts from the validated and verified models. This approach usually relies on many different Domain Specific Modeling Languages (DSML) either explicitly or through UML and its extensions that provides many different cooperating languages through diagrams (in fact, OMG is currently studying the possibility for the future next major

³ This work was funded by the European Union and the french DGCIS through the ARTEMIS Joint Undertaking inside the CESAR project
version of UML to define it as a collection of cooperating DSML) and profiles. Each DSML is defined as a specific metamodel or as an extension through profiles of a part of a huge metamodel in UML.

The use of components is also mandatory to improve the efficiency of system development. Common DSML do not usually integrate components natively, either because it was not an initial requirement, or to avoid a too complex definition of the language. Invasive Software Composition (ISC) was proposed by Aßman [1] in order to add a generic component structure to any existing DSML. This approach is the basis of ReuseWARE\(^4\) that provides ISC based tools inside the Eclipse Modeling Framework\(^5\). It allows to define the composition concern relying on elements in the metamodel and then to extract components from existing models with defined composition interface (called fragment boxes), and to compose fragments to produce new fragments or models. All the provided tools are generic and parametrized by the composition concern. The framework allows to adapt and extend an existing language by adding composition facilities at some points called Hook. This extension relies on a metamodel level transformation applied on the language definition based on the specification of the composition concern. The Hook are the variation points introduced in the models whose value can change and thus allows to build components. The main advantage of the ISC technology is that it is generic and can be applied to any language defined by a metamodel. This framework ensures that the result of the composition of fragments extracted from models conforming to a given metamodel is also conforming to the same metamodel. This common conformance is the kind of standard structural properties available in all the MDE frameworks that is verified in this paper. The long term purpose of our work is also to handle behavioral properties and thus tackle the formalization of all kind of compositional verification technologies.

In order to ease the integration of formal specification and verification technologies, some of the authors proposed in [23] a formal embedding of some key aspects of Model Driven Engineering in Set Theory. This embedding was then implemented using the Calculus of Inductive Construction and the Coq\(^6\) proof-assistant. This first version focused on the notions of models, metamodels, conformance and promotion. It was later extended to express constraints on metamodels using the Object Constraint Language (OCL). The purpose of this framework called Coq4MDE is to provide sound mathematical foundations for the study and the validation of MDE technologies. The choice of constructive logic and type theory as formal specification language allows to extract prototype tools from the executable specification that can be used to validate the specification itself with respect to external tools implementing the model driven engineering (for example, in the Eclipse Modeling Project).

This paper contributions are the specification of the composition operators provided by the ISC method [1] using an extension of Coq4MDE and the proof

\(^4\) http://www.reuseware.org
\(^5\) http://www.eclipse.org/modeling/emf
\(^6\) http://coq.inria.fr
of the well-foundedness and termination of these operators. This specification allows to express the models expected properties and the verification technologies for composite models and then provide support for compositional verification. This first contribution focuses on the metamodel structural conformance relation. It relies on the Model and MetaModel concepts from Coq4MDE that is extended to represent fragments as proposed by ISC. The various concepts provided by REUSEWARE are formalized leading to the proof that composition preserves metamodel conformity.

First, Section 2 introduces the notions of Model and MetaModel from Coq4MDE. Then, the REUSEWARE approach for extending DSML with components is presented in Section 3. The Coq4MDE framework is then extended to support the definition of component interface and the composition operators in Section 4. After that, the validation of a composition function is presented in Section 5. Also, a background of related work is given in Section 6. Finally, conclusion and perspectives are presented in Section 7.

2 Model and MetaModel

This section gives the main insight of our MDE framework Coq4MDE, derived from [23]. We first define the notions of model and metamodel. Then, we describe conformance using the conformsTo predicate.

Our approach separates the type level from the instance level, and describes them with different structures hence different types. A Model (M) is the instance level and a MetaModel (MM) is a modeling language used to define models (Figure 1). A MM also specifies the semantic properties of its models. For instance, in UML, a multiplicity is defined on relations to specify the allowed number of objects that have to be linked. Moreover, OCL is used to define more complex structural constraints which may not have any specific graphical notation.

Into our framework, the concept of MetaModel is not a specialization of Model. They are formally defined in the following way. Let us consider two sets: Classes, respectively References, represents the set of all possible class, respectively reference, labels. We also consider instances of such classes, the set Objects of object labels. References includes a specific inh label used to specify the inheritance relation. In the following text, we will withdraw the word label and directly talk about classes, references and objects.
Definition 1 (Model). Let $C \subseteq \text{Classes}$ be a set of classes.
Let $R \subseteq \{\langle c_1, r, c_2 \rangle \mid c_1, c_2 \in C, r \in \text{References} \}$ be the set of references among classes such that $\forall c_1 \in C, \forall r \in \text{References}, \text{card}\{c_2 \mid \langle c_1, r, c_2 \rangle \in R \} \leq 1$.

A model over $C$ and $R$, written $\langle MV, ME \rangle \in \text{Model}(C, R)$, is a multigraph built over a finite set $MV$ of typed object nodes and a finite set $ME$ of reference edges such that:

$$MV \subseteq \{(o, c) \mid o \in \text{Objects}, c \in C\}$$
$$ME \subseteq \{\langle\langle o_1, c_1, r, (o_2, c_2)\rangle\rangle \mid (o_1, c_1), (o_2, c_2) \in MV, \langle c_1, r, c_2 \rangle \in R \}$$

Note that, in case of inheritance, the same object label will be used several time in the same model graph, associated to different classes to build different nodes. This label reuse is related to inheritance polymorphism a key aspect of most OO languages. Inheritance is represented with a special reference called $\text{inh}$ (usually defined in the metamodeling languages such as MOF [17]).

Accordingly, we first define an auxiliary predicate stating that an object $o$ of type $c_1$ has a downcast duplicate of type $c_2$.

$$\text{hasSub}(o \in \text{Objects}, c_1, c_2 \in \text{Classes}, \langle MV, ME \rangle) \triangleq c_1 = c_2 \lor \exists c_3 \in \text{Classes}, \langle\langle o, c_2 \rangle, \text{inh}, (o, c_3)\rangle \in ME \land \text{hasSub}(o, c_1, c_3, \langle MV, ME \rangle)$$

Then, we define the notion of standard inheritance. The first part of the conjunction states that the inheritance relation only conveys duplicate objects. The second part states that every set of duplicates has a common base element (a common inherited class).

$$\text{standardInheritance}(\langle MV, ME \rangle) \triangleq \forall\langle\langle o_1, c_1, \text{inh}, (o_2, c_2)\rangle\rangle \in ME, o_1 = o_2 \land \forall\langle o_1, c_1 \rangle, \langle o_2, c_2 \rangle \in MV, o_1 = o_2 \Rightarrow \exists c \in \text{Classes}, \text{hasSub}(o_1, c_1, c, \langle MV, ME \rangle) \land \text{hasSub}(o_2, c_2, c, \langle MV, ME \rangle)$$

Finally, the following property states that $c_2$ is a direct subclass of $c_1$.

$$\text{subClass}(c_1, c_2 \in \text{Classes}, \langle MV, ME \rangle) \triangleq \forall \langle o, c \rangle \in MV, c = c_2 \Rightarrow \langle\langle o, c_2 \rangle, \text{inh}, (o, c_1)\rangle \in ME$$

Consequently, Abstract Classes, that are specified in the metamodel using the $\text{isAbstract}$ attribute, serve as parent classes and child classes are derived from them. They are not themselves suitable for instantiation. Abstract classes are often used to represent abstract concepts or entities. Features of an abstract class are then shared by a group of sibling sub-classes which may add new properties.

Therefore, a model does not conform to a metamodel if it contains objects that are instances of abstract classes without having instances of concrete derived classes as duplicates.

$$\text{isAbstract}(c_1 \in \text{Classes}, \langle MV, ME \rangle) \triangleq \forall \langle o, c \rangle \in MV, c = c_1 \Rightarrow \exists c_2 \in \text{Classes}, \langle\langle o, c_2 \rangle, \text{inh}, (o, c_1)\rangle \in ME$$

$\text{inh}$ must not be used in a model or metamodel as a simple reference.

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Definition 2 (MetaModel). A MetaModel is a multigraph representing classes and references as well as semantic properties over instantiation of classes and references. It is represented as a pair composed of a multigraph \((MMV, MME)\) built over a finite set \(MMV\) of class nodes and a finite set \(MME\) of edges tagged with references, and of a predicate over models representing the semantic properties.

A metamodel as a pair \(((MMV, MME), \text{conformsTo}) \in \text{MetaModel}\) such that:

\[
\begin{align*}
 MMV & \subseteq \text{Classes} \\
 MME & \subseteq \{(c_1, r, c_2) \mid c_1, c_2 \in MMV, r \in \text{References}\} \\
 \text{conformsTo} : \text{Model}(MMV, MME) & \rightarrow \text{Bool}
\end{align*}
\]

such that \(\forall c_1 \in MMV, \forall r \in \text{References}, \text{card}\{c_2 \mid (c_1, r, c_2) \in MME\} \leq 1\)

Given one model \(M\) and one metamodel \(MM\), we can check conformance. The \(\text{conformsTo}\) predicate embedded in \(MM\) achieves this goal. It identifies the set of valid models with respect to a metamodel.

In our framework, the conformance checks on the model \(M\) that:

1. every object \(o\) in \(M\) is the instance of a class \(C\) in \(MM\).
2. every link between two objects is such that there exists, in \(MM\), a reference between the two classes typing the two elements. In the following we will say that these links are instances of the reference between classes in \(MM\).
3. finally, every semantic property defined in \(MM\) is satisfied in \(M\). For instance, the multiplicity defined on references between concepts denotes a range of possible links between objects of these classes (i.e. concepts). Moreover, structural properties expressed on the metamodel as OCL constraints and behavioural properties will be taken into account in future work as \(\text{conformsTo}\) predicates.

This notion of conformity can be found in the framework depicted in Figure 1 by a dependency between a \(M\) and a \(MM\) it conforms to. In fact, the semantic properties associated to the metamodel are encoded into the \(\text{conformsTo}\) predicate. These semantic properties are not to be given a syntax. Instead, in order to express our properties, we assume an underlying logic that should encompass OCL in terms of expressive power.

In the rest of this paper, we extend the previous MDE framework to formalize compositional technologies. Our final target outside the scope of this paper is to formalize compositional verification activities. Coq4MDE is extended to support the introduction of components in DSML defined by their metamodels. This extension allows to express fragment boxes (models with defined interface) composition based on concepts from the ISC method.

In the scope of this paper, we take into account a simplified version of the \(\text{conformsTo}\) predicate (cf. Section 5) called \(\text{instanceOf}\) which is restricted to 1 and 2. We demonstrate that the verification of this \(\text{instanceOf}\) property is compositional relying on the ISC operators (the property of components is preserved in case of composition using the ISC basic operators).
3 ISC and ReuseWare approach

ISC [1] is a generic technology for extending a DSML with model composition facilities. Its first version was defined to compose Java programs and was implemented in the COMPOST system\(^8\). A universal extension called U-ISC was proposed in [12], this technique deals with textual components that can be described using context-free grammars and then the fragments are represented as trees. The method as presented considers tree merging for the composition. Recently, in order to deal with graphical languages the method was extended to support typed graphs in [14], this method was implemented in the ReuseWare framework. This last implementation is consistent with the description of models as graphs in our Coq4MDE framework.

ISC introduces the fragment box structure to group model or source code fragments. The fragment box defines its composition interface and then provides tools and concepts allowing the composition. The composition interface for a fragment box consists of a set of addressable points. Two types of addressable points are defined, the variation points which are elements inside the fragment box that can be used as a receptor for other elements and reference points which are used to address some parts inside a fragment box so they can be used in composition. We formalize thereafter one type of correspondence (variation/reference) points which is the pair (hook/prototype). As described in [11] a hook is a variation point that constitutes a place-holder to contain a fragment referenced by a prototype reference point.

We propose in the following section to extend the Coq4MDE framework to support ISC concepts and then to define a sound basis to ensure the correctness by construction for this composition style. This enables to describe and to verify structural properties. We plan in future work to extend the formalization to support other kind of properties and especially behavioural properties.

4 Formalizing Model Component Extraction and Composition

4.1 Extended MetaModel with Model Component

We must be able to extend any metamodel to support the definition of fragment boxes. This extension adds the definition of a fragment interface constituted from a set of addressable points. We note the extended metamodel for some metamodel \(MM\) as \(MM^{Ext}\). We note ROV the abstract class representing the addressable points, the Hook variation point and the Prototype reference points are subclasses of ROV. In \(MM^{Ext}\), every node in the graph representing \(MM\) can be referenced by an addressable point. For this purpose, an abstract class called AbSC is added as a super class for all the classes of \(MM\). This class is linked by the reference bind with ROV. The three classes ROV, Hook and

\(^8\) http://www.the-compost-system.org
Prototype are also automatically imported to the metamodel with appropriate inheritance relations between them.

The following definition represents the extension function implemented in Coq as a graph transformation which is not in the scope of this paper.

Definition 3. Let $MM = \langle\langle MMV, MME \rangle, \text{conformsTo} \rangle$ be a metamodel. Let $ROV, Hook, Prototype, AbsC \in \text{Classes}, \text{bind} \in \text{References}$. $MM^{Ext}$ is defined as $\langle\langle MMV^{Ext}, MME^{Ext} \rangle, \text{conformsTo}^{Ext} \rangle$ such that:

$$
\begin{align*}
MMV^{Ext} &= MMV \cup \{ROV, Hook, Prototype, AbsC\} \\
MME^{Ext} &= MME \cup \{(ROV, \text{bind}, AbsC)\} \\
\text{conformsTo}^{Ext}(\langle MV, ME \rangle) &\triangleq \text{conformsTo}(\langle MV, ME \rangle) \\
\land \text{isAbstract}(ROV) \\
\land \text{subClass}(Hook, ROV) \\
\land \text{subClass}(Prototype, ROV) \\
\land \text{isAbstract}(AbsC) \\
\land \forall c \in MMV, \text{subClass}(c, AbsC)
\end{align*}
$$

The figure 2 shows the example of the extension of the MetaModel $MM$.

4.2 Component interface extraction

The goal of the function FragmentExtraction is to construct a fragment box from a model by defining its composition interface. This function takes as parameters: a model, the object referenced in that model and the kind of the addressable point associated to this object.

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9 The metamodel extension used in [14] is defined at the third modeling level (metametamodel level) which may use the promotion notion to be defined in the Coq4MDE framework. The extension defined thereafter uses only the second modeling level (metamodel level) which seems to be sufficient.
FragmentExtraction : Model × Objects × Classes → Model is defined as:

\[
\text{FragmentExtraction}(⟨MV, ME⟩, o, HP) = ⟨MV^{Ext}, ME^{Ext}⟩
\]

where \( HP \in \{\text{Hook, Prototype}\} \) and \( \exists c \in \text{Classes}, (o, c) \in MV \) such that:

\[
\begin{align*}
MV^{Ext} &= MV \cup \{(h, HP), (h, ROV), (o, AbsC)\} \\
ME^{Ext} &= ME \cup \{(⟨(o, r), inh, (o, AbsC)⟩, \langle h, ROV⟩, bind, (o, AbsC)⟩, \langle h, HP⟩, inh, (h, ROV)⟩\}
\end{align*}
\]

ElimInterface eliminates the fragment box interface (all variation and reference points) of a fragment box, it is the inverse function of FragmentExtraction in case of only one addressable point in the fragment box. This is implemented in [14] using the remove operator which is automatically applied after composition execution to make the component understandable by tools where addressable points semantics is not defined.

\[
\text{ElimInterface} : \text{Model} \rightarrow \text{Model}, \text{such as:}
\]

\[
\text{ElimInterface} (⟨MV^{Ext}, ME^{Ext}⟩) = ⟨MV, ME⟩
\]

such that:

\[
\begin{align*}
MV &= \{(o, c) \in MV^{Ext}| c \notin \{\text{Hook, Prototype, VOR, AbsC}\}\} \\
ME &= \{(⟨(o, c), r, (o', c')⟩ \in ME^{Ext}| c, c' \notin \{\text{Hook, Prototype, ROV, AbsC}\}\}
\end{align*}
\]

The definition of these two functions requires some proofs on multigraphs. First, the proof that the extension of the multigraph representing the model is also a multigraph, this is done by proving that adding vertexes to a multigraph generates a multigraph and also adding edges in some conditions to a multigraph is also a multigraph. Second, the proof that deleting some elements from a multigraph representing the fragment box is also a multigraph, this is done using a filter function defined on multigraphs. So, CoQ4MDE can now support the definition of components with composition interface in any DSML. We describe in the following section the formalisation of ISC basics composition operators in CoQ4MDE.

4.3 Components Composition

In this section, we present the implementation in our framework of the two basic operators of ISC (bind and extend) presented in [1] [14]. The difference between these operators is that "the bind applied to the hook replaces the hook (i.e., it removes the hook from its containing fragment) while extend applied on a hook does not modify the hook itself but uses it as a position for extension (i.e., the hook remains in its containing fragment)."

---

10 Another version can be implemented by specifying a set of pairs \((o, HP)\) to add several points at the same time.


12 http://www.irit.fr/~Mounira.Kezadri/FISC/IntElim.html#elimInterface
**Bind** The bind operator replaces an object $o_1$ referenced by a hook variation point by an object $o_2$ referenced by a prototype reference point. The links to (resp. from) the object $o_1$ are replaced with links to (resp. from) the object $o_2$. The composed model is obtained by substituting the object $o_1$ by $o_2$ in both objects and links sets.\[bind : Model \times Model \times (Objects \times Classes) \rightarrow Model\]
is defined as:

$$
bind((MV_1, ME_1), (MV_2, ME_2), \langle b, B \rangle, \langle b', B' \rangle) = \langle MV_3, ME_3 \rangle
$$

where $\langle b, B \rangle \in MV_1$ and $\langle b', B' \rangle \in MV_2$, we have:

\[\forall h, p \in Objects, (\langle h, Hook \rangle, inh, (\langle h, ROV \rangle) \in ME_1 \]
\[\land (\langle b, B \rangle, inh, (\langle b, AbsC \rangle) \in ME_1 \]
\[\land (\langle p, Prototype \rangle, inh, (\langle p, ROV \rangle) \in ME_2 \]
\[\land (\langle p, ROV \rangle, bind, (\langle b', B' \rangle) \in ME_2 \]
\[\land (\langle b', B' \rangle, inh, (\langle b', AbsC \rangle) \in ME_2 \]

and finally:

$MV_3 = substV((\langle b, B \rangle, \langle b', B' \rangle), MV_1)$

$ME_3 = substE((\langle b, B \rangle, \langle b', B' \rangle), ME_1)$

such that $substV((\langle b, B \rangle, \langle b', B' \rangle), MV)$ (resp. $substE((\langle b, B \rangle, \langle b', B' \rangle), ME)$) is the function that replaces $(b, B)$ by $(b', B')$ in every element in $MV$ (resp. relation in $ME$). The condition of the composition is: $B = B'$.

The construction of this function in Coq requires the proof that substituting an object by another in some multigraph is also a multigraph.\[13\] The proof is done by induction, it is automatic for the empty graph. In case of a graph built from adding an edge (a reference) to the graph, one reference is presented as $\langle src, dst, a \rangle$, suppose that the substitution replaces $o_1$ by $o_2$, we must consider all cases of equality between $src$, $dst$, $o_1$ and $o_2$. Last, in case of a graph built by adding a vertex to a graph which considers also cases of equality between the added vertex, $o_1$ and $o_2$. The current implementation can be largely improved by the definition of some graph operations like the map function, which is currently partially done and will be presented in future work. A recursive call for the previous function using a list of correspondence (Variation/Reference) points allows to replaces several objects at the same time.

**Extend** This operator allows to extend a model $(MV_1, ME_1)$ (the extension point is an object $o_1$ addressed as a hook variation point inside the model) by a model $(MV_2, ME_2)$ at an object $o_2$ addressed as a prototype reference point.

This function is parametrized by a metamodel (to insure the type safety) and a name for the added link between $o_1$ and $o_2$. The composed model consists of a multigraph built over the union of all objects of $(MV_1, ME_1)$ and $(MV_2, ME_2)$, all links of the two models in addition to a link between the objects $o_1$ and $o_2$.

\[extend : Model \times Model \times (Objects \times Classes) \times (Objects \times Classes) \times MetaModel \times References \rightarrow Model\]
is defined as:

\[http://www.irit.fr/~Mounira.Kezadri/FISC/CompBind.html#GraphSubst\]
extend(⟨MV₁, ME₁⟩, ⟨MV₂, ME₂⟩, ⟨b, B⟩, ⟨b′, B′⟩,
(⟨MMV, MME⟩, conformsTo), LinkName) = ⟨MV₃, ME₃⟩
where ∃⟨b, B⟩ ∈ MV₁ and ⟨b′, B′⟩ ∈ MV₂, we have :
extensible(⟨MV₁, ME₁⟩, ⟨MV₂, ME₂⟩, ⟨b, B⟩, ⟨b′, B′⟩,
(⟨MMV, MME⟩, conformsTo), LinkName) such that :
MV₃ = MV₁ ∪ MV₂
ME₃ = ME₁ ∪ ME₂ ∪ {⟨⟩(⟨b, B⟩, LinkName, ⟨b′, B′⟩)}

The predicate extensible checks that a model ⟨MV₁, ME₁⟩ whose interface is ⟨b, B⟩ regarding some metamodel can be extended by another model ⟨MV₂, ME₂⟩ whose interface is ⟨b′, B′⟩.

The predicate isExtendedH verifies that ⟨b, B⟩ is a hook in ⟨MV₁, ME₁⟩.

isExtendedH(MV₁, ME₁)⟨b, B⟩ ≜
∃h ∈ Objects, ⟨⟨h, Hook⟩, inh, ⟨h, ROV⟩⟩ ∈ ME₁
∧⟨⟨h, ROV⟩, bind, ⟨b, AbsC⟩⟩ ∈ ME₁
∧⟨⟨b, B⟩, inh, ⟨b, AbsC⟩⟩ ∈ ME₁

The predicate isExtendedP verifies that ⟨b, B⟩ is a prototype in the model.

isExtendedP(MV₂, ME₂)⟨b, B⟩ ≜
∃p, ⟨⟨p, Prototype⟩, inh, ⟨p, ROV⟩⟩ ∈ ME₂
∧⟨⟨p, ROV⟩, bind, ⟨b, AbsC⟩⟩ ∈ ME₂
∧⟨⟨b, B⟩, inh, ⟨b, AbsC⟩⟩ ∈ ME₂

The construction of this function in Coq requires the proof that the multigraph built by extending another multigraph as described in the function extend is also a multigraph.¹⁴

Here we defined only one type of correspondence variation and reference point (hook/prototype), the method as presented in [14] considers also another type of correspondence (slot/anchor). The second type requires to consider the containment property of an edge. The difference as explained in [14] is that contrarily to hook and prototype the slot variation point and the anchor reference point keeps their containments in case of composition. The first type of correspondence allows to express quite complicated composition functions like described in the following example and is consistent with the current models graph representation. The second type of correspondence can be considered in future work. The operators like described here are applied to the two models, a generalization to

¹⁴ http://www.irit.fr/~Mounira.Kezadri/FISC/CompBind.html#compositionExtend
an application on several models at the same time is allowed in ReuseWare and can be implemented in our framework as an iterative application of the operators by composing the models one by one or by defining more general operators that can be applied on several models.

4.4 Detailed example

We describe in this section the use of the previously defined basic operators to elaborate a model composition. $M_1$ is a state machine modeling a door with a lock. The door provides the operations: open, close, pass, lock and unlock. We would like to add the possibility of simple and double locking the door, these two states are described in the model $M_2$. $M_1$ and $M_2$ are described in Fig. 3.

The first step is to define the interface for each model. This is done with the FragmentExtraction function, the function applied to the model $M_1$ defines Locked as a hook and applied to $M_2$ defines Simple lock as a prototype like described in Fig. 4.

The application of the function bind on the two fragments as described in Fig. 4 followed by the elimination of the interface produces the model $M_{bind}$ shown in Fig. 5.
Then, Simple lock is defined in $M_{bind}$ as a prototype reference point and Double lock is defined in $M_{2,fp,elim}$ as a hook variation point as shown in Fig. 6.

![Fig. 6. Fragment boxes extraction](image1)

The execution of the function $extend$ on the two models in Fig. 6 after the interface elimination generates the model presented in Fig. 7. The model is the state machine for a door with a double lock option.

![Fig. 7. Model after execution of the extend and ElimInterface functions](image2)

The original contribution of this paper is not the definition of composition operators which is taken from ISC but their implementation in the Coq proof assistant, their integration in the Coq4MDE framework and the proof that the verification of the $instanceOf$ property is compositional with respect to these operators.

5 Composition Validation

The $bind$ and $extend$ operators are defined in order to enforce the well typedness properties. These two operators like all the concepts presented in this paper are encoded in the Coq proof assistant. The aim of this formalization is to check some properties on the composite models and then provide the basis for the specification and proof of correctness of compositional verification technologies. The first property considered is the well typedness property. This property is related to the conformance defined in Section 2. It checks that every object in $M$ is the instance of a class in $MM$ and every link in $M$ is an instance of a relation in $MM$. To prove that this verification is compositional, we need to prove that the composition of two models instances of the same metamodel is also an instance of the same metamodel.

We define the first validity criteria for any composition function. This criteria is defined as a higher order predicate that checks the well typedness for some function. The function $instanceOf$ is used in that purpose, it checks that
all objects and links of a Model are instances of classes and references in a metamodel.

$$\text{InstanceOf}((\langle\langle MV, ME \rangle, \langle\langle MMV, MME, \text{conformsTo} \rangle\rangle)) \triangleq \forall \langle o, c \rangle \in MV, c \in MMV \land \forall \langle\langle o, c, r, \langle o', c' \rangle \rangle \in ME \land \langle c, r, c' \rangle \in MME$$

Then, the predicate ValidCompositionFunction$_{MM}$ reflects this criteria. It verifies that using two components instance of $MM$, the component resulting from the application of a composition function $f$ is also instance of $MM$.

$$\text{ValidCompositionFunction}(MM \in \text{MetaModel}, f) \triangleq \forall M1 M2 \in \text{Model}, \text{InstanceOf} (M1, MM) \land \text{InstanceOf} (M2, MM) \rightarrow \text{InstanceOf} ((f M1 M2), MM)$$

We use this predicate to verify the type safety for the composition operator $\text{bind}$ described in Section 4.3. This is described in the theorem ValidBind.

Theorem ValidBind : $\forall MM \in \text{MetaModel}$,
$$\text{ValidCompositionFunction}(MM, \text{bind})$$

The Coq proof is done for this theorem. It uses intermediate lemmas that prove the preservation of the well typedness by the elementary operations implied in the composition. Among these lemmas, conformsAddO ensures that the result of adding an object instance of a class in the metamodel to a component instance of this metamodel is a component instance of the same metamodel.

Theorem conformsAddO :
$$\forall (MV, ME) \in \text{Model}, ((MMV, MME, \text{conformsTo}) \in \text{MetaModel}. \forall o \in \text{Objects}, c \in \text{Classes}.
\text{InstanceOf}((\langle MV, ME \rangle, ((MMV, MME, \text{conformsTo})) \land c \in MMV
\rightarrow \text{InstanceOf}((\langle MV \cup \{\langle o, c \rangle\}, ME, ((MMV, MME, \text{conformsTo}))$$

Another Coq proof was done to demonstrate the type safety for the composition operator $\text{extend}$ described also in Section 4.3. This is encoded in the theorem ValidExtend.

Theorem ValidExtend : $\forall MM \in \text{MetaModel}$,
$$\text{ValidCompositionFunction}(MM, \text{extend})$$

Also, similar correctness properties should hold for the fragment extraction function and the elimination function.

Theorem ValidFragmentExtraction :
$$\forall (MV, ME) \in \text{Model}, ((MMV, MME, \text{conformsTo}) \in \text{MetaModel}.
\forall o \in \text{Objects}, HP \in \{\text{Hook, Prototype}\}.
\text{InstanceOf}((\langle MV, ME \rangle, ((MMV, MME, \text{conformsTo}))
\rightarrow \text{InstanceOf}(\text{FragmentExtraction}(\langle MV, ME \rangle, o, HP), (\langle (MMV^{Ext}, MME^{Ext}), \text{conformsTo}^{Ext}))$$
Theorem $\text{ValidInterfaceElimination}$:
\[
\forall (M_V, M_E) \in \text{Model}, ((M_{MV}, M_{ME}), \text{conformsTo}) \in \text{MetaModel}.
\]
\[
\text{InstanceOf}((M_V, M_E), ((M_{MV}^{\text{Ext}}, M_{ME}^{\text{Ext}}), \text{conformsTo}^{\text{Ext}}))
\]
\[
\rightarrow \text{InstanceOf}(\text{InterfaceElimination}((M_V, M_E)),
\]
\[
((M_{MV}, M_{ME}), \text{conformsTo}))
\]

So, starting from the Coq4MDE framework and from the ISC composition method, we defined a framework for model composition. The definitions of model and metamodel were extended to support the definition of model composition interface, the constituted fragment box is also a model conforms to an extended metamodel. The basic composition operators was described like all elements in this paper using the Coq proof assistant. The source code is about 6400 lines, it is accessible at http://www.irit.fr/~Mounira.Kezadri/FISC/index.html. The formalization in Coq ensures the termination of the composition operators, elaborates a compositional verification property and also will enable to describe and prove more richer properties in future work.

6 Related work

6.1 Composition approaches

Models are aspects of the system that must be composed to build the final system, similarly to aspects in AOP [15]. Tools and approaches have been proposed aiming to automate the composition task. This problem concerns a wide variety of modeling domains and includes several techniques. We are looking for an approach that supports component extraction from models and model composition from components. The ISC approach supports these two characteristics. It enables to extend arbitrary language to provide reuse with the concepts of fragment box. In this method components can be invasively composed, this can be done by adapting or extending the component at some variation point (fragments or positions, which are subject to change) by transformation. Several composition methods were collected in [13]. most of these methods are interested in implementing the merge operator by using some mappings between the models like Rational Software Architect 16, Bernstein et al. data model [5], Atlas Model Weaver 17 [9], Epsilon 18, Theme/UML [7] and EMF Facet19. Merge operators as presented in these works can be implemented in our framework and constitutes one of the directions for future work.

---

15 We can’t write any function in Coq if the proof of termination is not given or deduced by Coq
17 http://www.eclipse.org/gmt/amw/
18 http://www.eclipse.org/gmt/epsilon/
19 www.eclipse.org/proposals/emf-facet/
6.2 Formalization of model driven engineering

MoM ENT (MOdel manageMENT) [6] is a model management framework based on experiments in formal model transformation and data migration, it provides a set of generic operators to manipulate models. MoM ENT relies on algebraic formalisms using the Maude language [8]. In this framework, the metamodels are represented as algebraic specifications and the operators are defined independently of the metamodel. To be used, the operators must be specified in a module called signature that specify the constructs of the metamodel. The approach was implemented in a tool 20 that gives also an automatic translation from an EMF metamodel to a signature model.

A. Vallecillo et al. have designed and implemented previously a different embedding of metamodels, models ([22]) and model transformations ([24]) using MAUDE. This embedding is shallow, it relies strongly on the object structure proposed by MAUDE in order to define model elements as objects, and relies on the object rewriting semantics in order to implement model transformations.

I. Poernomo has proposed an encoding of metamodels and models using type theory ([19]) in order to allow correct by construction development of model transformation using proof assistant like Coq ([20]). Some simple experiments have been conducted using Coq mainly on tree-shaped models ([21]) using inductive types. General graph model structure can be encoded using co-inductive types. General graph model structure can be encoded using co-inductive types. General graph model structure can be encoded using co-inductive types. General graph model structure can be encoded using co-inductive types. However, as shown in [18] by C. Picard and R. Matthes, the encoding is quite complex as Coq enforces structural constraints when combining inductive and co-inductive types that forbid the use of the most natural encodings proposed by Poernomo et al. M. Giorgino et al. rely in [10] on a spanning tree of the graph combined with additional links to overcome that constraint using the ISABELLE proof assistant. This allows to develop a model transformation relying on slightly adapted inductive proofs and then extract classical imperative implementations. These embeddings are all shallow: they rely on sophisticated similar data structure to represent model elements and metamodels (e.g. Coq (co-)inductive data types for model elements and object and (co-)inductive types for metamodel elements).

The work described in this paper is a deep embedding, each concept from models and metamodels are encoded using elementary constructs instead of relying on similar elements in MAUDE, Coq or ISABELLE. The purpose of this contribution is not to implement model transformation using correct-by-construction tools but to give a kind of denotational semantics for model driven engineering concepts that should provide a deeper understanding and allow the formal validation of the various implemented technologies.

6.3 Formalization of models composition

A formalisation of ISC in Frame Logic (or F-Logic) [16] was proposed in [2]. F-Logic provides structural aspects of object oriented and frame-based languages (object identity, complex objects, inheritance, polymorphic types, query

20 http://moment.dsic.upv.es/
methods, encapsulation and others). The description in F-Logic allows reasoning on the composition architecture and provides many additional checking: cyclic check, reachability and constraint check. In this work we define the mathematical formalisation of the concepts of the ISC method aiming to describe it in a proof assistant. The advantage of this formalization in addition to those of the previous cited work (it can be added to any model and is independent of specific component description languages and the checked properties), are proof of termination of composition functions and the possibility of extracting the validated executable code from the definitions after some modifications on functions that are written now for validation purpose.

6.4 Compositional verification

In order to develop safety critical systems, methods are now needed that allows not only the reuse of components but also of their properties for inferring the global properties of the composite system from properties of his constituent components. Nguyen, T.H. proposes in [4] a compositional verification approach to check safety properties of component-based systems. The systems must be described in the BIP (Behavior - Interaction - Priority) language [3]. Another approach allowing to verify systems by composition from verified components was proposed in [25], this approach reduces the complexity of verifying component-based systems by utilizing their compositional structures. In this approach, temporal properties of a software component are specified, verified, and packaged with the component. The selection of a component for reuse considers also its temporal properties. The Ptolemy$^{21}$ project proposes a compositional theory for concurrent, real-time, embedded systems. It uses well defined models of computation and defines an unified mathematical framework to relate heterogeneous models of computation. In this paper, regarding the previous cited methods, we adopted a generic composition technology where the interactions and temporal properties are not yet integrated. This is planned for future work.

7 Conclusion

Starting from Coq4MDE our formal framework for model and metamodel definition, we have tackled the problem of model composition. Taking inspiration from the ISC generic method for model composition and also from the Reuse-Ware toolbox, we proposed first a metamodel extension, and associated model operators for expressing component extraction and composition. This yielded a formalisation of model components, model extraction and model composition. All these notions are also currently being reflected in the Coq proof assistant, following the line of thought of our previous work around model and metamodel formalisation. This embedding provides us correct-by-construction pieces of executable code for the different model operations related to composition. For

$^{21}$http://ptolemy.eecs.berkeley.edu/
instance model extraction and model composition are both proved to be terminating, the latter operation being in addition correct, as advocated by the main theorem. As we target a general purpose MDE-oriented framework, our work applies to any model, modeling language, application and is not restricted to some more-or-less implicit language context.

Yet, for the ease of experimentation, we have in a first step somehow restricted the possibilities of our composition framework. For instance, the notion of conformity, a notion at the heart of our formal description, has been temporarily weakened to take into account only instantiation constraints, disregarding any other model property (multiplicity, etc).

As future work, all these constraints should be enforced to achieve a fully-fledged formal model composition framework.

Furthermore, the interplay between model composition (where objects are replaced by others, assuming they have the same type) and sub-typing (where a single object may exhibit many types, due to duplication) needs to be clearly worked out in our framework.

This proposal is a preliminary mandatory step in the formalization of compositional formal verification technologies. We have tackled the formal composition of models from model fragments independently of the properties satisfied by the model fragments and the expected properties for the composite model. The next step in our work is to formalize the notion of model verification relying on several use case from simple static constraints such as typing or verification of OCL constraints satisfaction, to more dynamic properties such as deadlock freedom as proposed in the BIP framework. The expected result of our work is a framework to define compositional verification technologies and to prove the correctness of the associated verification tools.

References

Controlling an iteration-wise coherence in dataflow

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Abstract. This paper formalizes a data-flow component model specifically designed for building real-time interactive scientific visualization applications. The advantages sought in this model are performance, coherence and application design assistance. The core of the article deals with the interpretation of a property and constraint based user specification to generate a concrete assembly based on our component model. To fulfill one or many coherence constraints simultaneously, the application graph is processed, particularly to find the optimal locations of filtering objects called regulators. The automatic selection and inter-connection of connectors in order to maintain the requested coherences and the highest performance possible is also part of the process.

Keywords: Composition, Coherence, Coordination, Synchronization

1 Introduction

Assisted or semi-automated composition is a recurrent feature in component-based frameworks [6], particularly when the end users are not computer scientists. The aim is to provide an abstraction layer that makes composition more intuitive, descriptive and, ideally, close to the natural language. Research in this area addresses the underlying reasoning approaches that would map the user’s specification to the concrete assembly of the model’s elements. Apart from hiding the technicalities of the model, the purpose of allowing a coarse grained specification is to alleviate the complexity of tuning a whole system, a complexity that grows exponentially with the size of this system.

Automation can take place in two aspects of dataflow composition: consistency and coordination. The former consists in ensuring the compatibility of the data exchanged by the components and is an inescapable feature for scientific workflow designers [2,13]. The latter deals with the execution order of the components. In models where connection patterns are mainly blocking, i.e. synchronous, the execution of the components is sequential. Solutions have then to be provided to allow users to put loops or branching in their workflows so that they can accurately set up their processing scenarios. While some approaches [4,7,11,14] propose ready-to use control Constructs, others [3] suggest

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composition languages to build advanced coordination patterns out of simpler ones.

The level of abstraction of the application’s specification that the different approaches propose closely depends on the targeted audience and application areas. The component approach has, for example, been widely used in scientific computing. A variety of Scientific Workflow Management Systems (SWMSs) [18] exist to design, generate, deploy and execute scientific applications. The targeted applications usually consist in carrying out an overall process over a dataset through a sequence of finite steps. Despite the name “workflow”, the current state of the art of SWMS is divided into frameworks adopting either a workflow paradigm [5] or a dataflow paradigm [7, 11, 14]. Because they are less dependent on the components’ implementations -no function calls between components, only data is exchanged-, dataflow-oriented frameworks promote code reuse better. In SWMSs, the trend is to bring the specification to an always higher level.

In [12], the authors suggest to refine the results of a workflow execution with intents and goals expressed at specification. In SWMSs also, the processing pipeline that produced a result is referred to as the provenance of this result [15] and is a crucial information for scientists. Provenance is usually recorded and displayed at the end of an execution for analysis [1, 11] or for failure diagnosis [17]. However, because it is itself seen as part of the result set, the SWMSs do not provide any interface for a priori controlling or parameterizing provenance. This would though help ensure the accuracy of a result depending on the coherence of its different sources.

In [10], we introduced a component model specifically designed for high performance interactive scientific applications. In that model, components can encapsulate different kinds of tasks: computing, display, user control management, data conversion, etc. They, by definition, run iteratively and their composition is the loosest possible to promote performance. It was presented along with a coarse grained coordination specification system. Coupling is usually loose in such applications so branching control is not necessary and coordination rather defines the degree of synchronicity between components. Nevertheless, spatial and temporal provenance remain important. That is why, in our model’s specification system, we introduced the possibility of imposing tight coherence constraints which consisted in allowing the user to request an exact synchronicity between message flows reaching the same component. This property is, to scientists, among the relevant information [16] when evaluating their results. Our contribution was then to automatically adapt the user’s initial graph to fulfill this type of constraints. In the current paper, we intend to enrich the definition of coherence and the component model to allow looser user provenance constraints.

This paper is organized as follows: Section 2 introduces our component model. Section 3 details our methodology to automatically build a coherent dataflow out of a user specification. In Section 4, we evaluate our method and give the axes of our future work.
2 Component model

In [10], we have defined a component model for Real-time interactive (RTI) applications including a component of iterative nature and five inter-component connection patterns. We also described how this model and our connection patterns can be used to construct an application guaranteeing a tight coherence of the data consumed by a component. In this section, we briefly give a reminder of our model and add to it a new object called the regulator.

2.1 Components

A component works iteratively. It is defined as a quadruple $C = (n, I, O, f)$ where $n$ is the name of the component and $I$ and $O$ two sets of user defined input and output ports. $I$ and $O$ respectively include $s$ (for start) and $e$ (for end), two default triggering input and output ports. $f$ is a boolean to indicate that the component must run freely and that its iteration cycle can not be blocked by other components. The iteration cycle of the component consists in

1. receiving new messages on all its connected input ports, including $s$,
2. when all its input ports are fed, beginning a new iteration,
3. at the end of the iteration, producing new data on all its output ports and an ending signal on $e$ that can be connected to the $s$ port of another object to trigger it.

Each component numbers its iterations. input and output ports are identified by a name and data circulating between ports are called messages. Along with the data it transports, a message $m$ also contains stamps. A stamp is a small information associated to a message and generated by the sender. Each message contains at least one stamp, denoted $it(m)$, that is the iteration number of the component that produced it. The components of our model can also handle empty messages, i.e. containing no data, allowing it to go out of the waiting state as soon as all of its input ports are supplied. For a component $C$, $name(C)$ denotes its name and $I(C)$ and $O(C)$ respectively its sets of input and output ports. A port of a component $C$ is denoted $C.i$ with $i \in I(C)$ or $C.o$ with $o \in O(C)$.

2.2 Connectors

Connectors must be set between two components to determine the communication policy between them, i.e. the type of synchronization and the possibility to lose messages or not. A connector $c$ is a quadruple $c = (n, \{s, i\}, \{o\}, t)$ where $t$ is its type and $i$ is an input port and $o$ an output port. $n$ and $s$ are similar to their homonyms in the component. We use the same notations $name(c)$ and $type(c)$ as for components. $c$ can store several messages. When the sender writes a message on an output port, it simply adds this message to the connector and when the receiver reads its input ports, the connector delivers one of its messages.
Because the components might run at different rates, the connectors need to avoid the overflow of messages when the receiver is slower than the sender. On the other hand, the sender might also slow the receiver down if its iteration rate is lower. To tackle these problems, we propose five connection patterns besides the plain FIFO, summarized in Figure 1. These connectors needed to be carefully designed in order to express fine inter-components synchronization policies.

- The sFIFO connector is a plain FIFO connection where, to prevent overflows, the sender waits for a triggering signal on its s port usually sent by the receiver.
- The bBuffer and nbBuffer keep their incoming messages until the reception of a triggering signal and then dispatches the oldest message. These buffered FIFO connections can be useful to absorb overflows when one of the two components has an irregular iteration rate. The n(on)b(locking)Buffer connector dispatches empty messages when its buffer is empty whereas the b(locking)Buffer blocks the receiver until fresh messages arrive.
- A greedy connector keeps only the last message provided by the sender and sends it upon the receiver’s request. It is usually used to avoid overflows when it is not required that all the messages are processed. The bGreedy and the nbGreedy are, respectively, the blocking and the non-blocking variants of this pattern.

2.3 Regulators

Regulators are special multi-channel connectors that coordinate the message flows of several communication channels. Their policy is expressed by user-defined coherence rules to filter the message flows on the different channels. These rules are linear formulae over message iteration numbers. Formally, a regulator, illustrated in Figure 2, is a quintuple $r = (n, I, O, F, b)$ where $n$ is its...
name, \( I \) and \( O \) its sets of input and output ports. \( I \) contains a triggering port \( s \). \( F \) is a set of formula, also denoted \( F(r) \). A formula has the form \( \text{in}_i \circ \alpha \times \text{in}_j + \delta \) with \( \text{in}_i, \text{in}_j \in I \setminus \{s\}, \circ \in \{\leq, =, \approx\} \) and \( \alpha, \delta \in \mathbb{N} \). The operator \( \approx \), used with \( \delta > 0 \), denotes an absolute gap tolerance of \( \delta \) between the two operands \( \text{in}_i \) and \( \text{in}_j \). \( b \) is a boolean that denotes the blocking behaviour of the regulator. Moreover there is a one to one correspondence between the ports of \( I \setminus \{s\} \) and those of \( O \). These two sets thus contain the same number of ports.

Let \( M = \{m_1, \ldots, m_n\} \) be a set of messages contained in each buffer of messages received by \( r \) on its \( n \) input ports. We say that \( M \) validates \( f = \text{in}_i \circ \alpha \times \text{in}_j + \delta \) of \( F(r) \) if \( \text{it}(m_i) \circ \alpha \times \text{it}(m_j) + \delta \). \( M \) validates \( F(r) \) if it validates all the formulae of \( F(r) \).

The behaviour of a regulator is the following:

1. it buffers the messages received on its input ports,
2. each time it receives a signal on its port \( s \), it analyzes the iteration numbers of the messages available in its input buffers,
3. (a) if a set of messages that validates \( F(r) \) can be found in the buffers, the regulator moves them to the corresponding output ports and flushes the older messages in the buffers. Besides, if, in an input buffer, more than one messages fulfills the rules, the oldest one is selected.
   (b) otherwise, the regulator dispatches empty messages from all of its output ports if \( b \) is set to false and does nothing if not.

Thanks to blocking connectors or to the synchronization mechanisms described in Section 3, the coherence established by a regulator can be maintained throughout the application.

\begin{figure}[h]
\centering
\includegraphics[width=0.5\textwidth]{fig2}
\caption{Schema of the regulator}
\end{figure}

2.4 Links

Links connect components, connectors and regulators together via their ports. They are denoted by \((x.p, y.q)\) with \( x, y \) components, connectors or regulators, \( p \in O(x) \) and \( q \in I(y) \). There are two types of links:

- A data link transmits data messages. For a data link \((x.p, y.q)\), we impose that \( p \neq e, q \neq s \) and at least \( x \) or \( y \) is a connector or a regulator. Indeed,
as a connector or a regulator is always required to define a communication policy, a data link cannot be directly set between two components.

- A triggering link transmits triggering signals. For such a link \((x.p, y.q)\), we impose that \(x\) is a component, \(p = e\) and \(q = s\). Please note that, to avoid deadlocks, neither components nor connectors nor regulators wait for a triggering signal before their very first iteration.

### 2.5 Application graph

With these elements, an application is represented by a graph called the application graph. The vertices of this graph are the components, the connectors and the regulators. The edges represent the links.

**Definition 1.** Let \(C\) be a set of components, \(L\) a set of connectors, \(R\) a set of regulators, \(D\) a set of data links, \(T\) a set of triggering links. The graph \(G = (C \cup L \cup R, D \cup T)\) defines an application graph. In the remainder of this article, we call a data path of \(G\) an acyclic path in the graph \((C \cup L \cup R, D)\).

With \(G\) an application graph, let us also consider the following additional definitions:

- We call the source \(src(p)\) the starting vertex of a data path \(p\) of \(G\) and destination \(dest(p)\) its ending vertex.
- A message \(m\) arriving at \(dest(p)\) is called a result of \(p\) and the message from the source that originates this result is denoted by \(ori(m)\).
- A data path whose source and destination are components is called a pipeline.
- \(rank_p(x)\) denotes the rank of element \(x\) along pipeline \(p\). \(\text{rank}_p(src(p)) = 1, \text{rank}_p(dest(p)) = \text{length}(p)\) with \(\text{length}(p)\) the number of elements of \(p\).

Figure 3 illustrates a sample application graph.

### 3 Provenance-based coherence

This section describes a composition method to build an application that can be deployed on a distributed architecture. We aim to propose an automatic process in a few steps to transform a specification graph defined by a scientist into an application graph respecting all the coherence constraints and allowing the best performance possible.

#### 3.1 Specification graph

Application specification helps the user focus on the expected properties of the communications in the application, sparing him technicalities. It is done through a directed graph called the specification graph. The vertices of this graph are the components of the application and its edges indicate which component ports are connected together. Its vertices are the components defined in Section 2.1. The edges, directed from the sender to the receiver, are labelled with the output and input ports and with constraints on the communications. These constraints are of two types.
the message policy, i.e. can this communication drop messages or not,
the synchronization policy, i.e. should the receiver of the message be blocked
when no new messages are available.

These communication constraints are used to construct a preliminary application graph where connectors are automatically chosen to implement the synchronization policy with the best performance possible but without any guarantee on coherence. Besides the graph itself, a set of constraints $\mathcal{K}$ defines the coherence constraints on the input ports of the components. Provenance-based coherence is a fine type of coherence based on the tolerated -positive, null or negative- iteration gap between two messages $m_1$ and $m_2$ issued by two output ports, and originating the messages that arrive simultaneously to two input ports of a component at each iteration of it. While in [10] we introduced a tight coherence imposing equalities between message iterations and a common component as message source, this new coherence type aims at allowing more flexible synchronization policies when the application needs not to manipulate data generated exactly at the same iteration by the same component. More formally provenance-based coherence is defined as follows:
Definition 2. Let $C_1, C_2$ and $D$ be three components such that $C_1 \neq D, C_2 \neq D$, $o_i \in O(C_1), o_j \in O(C_2)$ and $\{i_k, i_l\} \subset I(D)$. The coherence constraint $\kappa$ defined by $D_{i_k, i_l} : C_1.o_i \circ \alpha \times C_2.o_j + \delta$ with $\circ \in \{\leq, =, \approx\}$ and $\{\alpha, \delta\} \in \mathbb{N}$ is satisfied if, for each pair of pipelines $p_1$ and $p_2$ starting respectively at $C_1.o_i$ and $C_2.o_j$ and reaching respectively $D.i_k$ and $D.i_l$, we ensure that $\text{it}(\text{ori}(m_1)) \circ \alpha \times \text{it}(\text{ori}(m_2)) + \delta$ where $m_1$ and $m_2$ are results of respectively $p_1$ and $p_2$ read at the same iteration of $D$. Such a pair of pipelines $p_1$ and $p_2$ are called sibling pipelines with respect to coherence $\kappa$. $\text{sib}_\kappa(p)$ denotes the set of sibling pipelines of pipeline $p$ with respect to coherence $\kappa$.

Figure 4 gives an example of specification graph to which we add the following provenance coherence constraints:

- $\kappa_1 = J_{in_1, in_2} : A.out \approx B.out + 10$, which means that, at each iteration of component $J$, we do not allow the pair of messages read on $in_1$ and $in_2$ of $J$ to reflect an absolute iteration difference between $A$ and $B$ that is greater than 10 iterations.
- $\kappa_2 = K_{in_1, in_2} : E.out \approx C.out + 5$, which has the same meaning as the previous constraint.

Fig. 4. A specification graph

3.2 Preliminary application graph

The first step of the process consists in building a preliminary application graph by replacing each edge of the specification graph with a connector following the rules of Table 1. As in many cases several connectors fit the same combination,
this table was created following the rule: *The generated application has to be, first of all, as overflow-safe as possible and then, as fast as possible.*

The application graph of Figure 3 derives from the specification graph of Figure 4. Of course, if no provenance coherence is requested, the application graph can be finalized just after this step.

### 3.3 Coherence subgraphs

The next steps of the process deal with the solving of the coherence constraints. The first step of the transformation consists in looking, in the application graph, for pipelines that must be coherent. They are collected into coherence subgraphs.

**Definition 3.** Given an application graph $G$ and $C_1, C_2$ and $D$ three distinct components of $G$ such that $o_i \in O(C_1), o_j \in O(C_2)$ and $\{i_k, i_l\} \subset I(D)$ and given a coherence constraint $\kappa = D_{i_k, i_l} : C_1.o_i \circ \alpha \times C_2.o_j + \delta$, the coherence subgraph $g_\kappa$ of $\kappa$ is the subgraph of $G$ that contains all the sibling pipelines between the source ports $C_1.o_i$ and $C_2.o_j$ and the destination ports respectively $D.i_k$ and $D.i_l$.

The coherence subgraphs of $\kappa_1$ and $\kappa_2$ are in respectively a dashed and a dotted frame in Figure 3. As they intersect, they are merged into one single subgraph to avoid backtracking in the remaining of the process.

In a subgraph, we can decompose each path into a set of independent synchronous segments according to the following definition.

**Definition 4.** A pipeline $(C_1, c_1, \ldots, C_{n-1}, c_{n-1}, C_n)$ where $C_i$ ($1 \leq i \leq n$) is a component and $c_i$ ($1 \leq i \leq n - 1$) is either a sFIFO or bBuffer connector is called a synchronous segment.

The message flow is preserved inside a synchronous segment i.e. no messages are lost and no empty messages are produced by the connectors. As a consequence, all the components of the segment perform the same number of iterations.

**Property 1.** Let $s = (C_1, c_1, \ldots, C_{n-1}, c_{n-1}, C_n)$ be a synchronous segment and $m_n$ a message produced by $C_n$, $it(m_n) = it(ori_k(m_n))$.

The property is obvious since no message is lost inside a synchronous segment. $C_n$ generates as many messages as $C_1$. 

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<table>
<thead>
<tr>
<th>Blocking policy</th>
<th>Non-blocking or Free receiver</th>
</tr>
</thead>
<tbody>
<tr>
<td>Msg loss</td>
<td>bGreedy</td>
</tr>
<tr>
<td>Free sender</td>
<td>sFIFO</td>
</tr>
</tbody>
</table>

Table 1. The communication pattern selection
Definition 5. The connector between two successive synchronous segments is called a junction and is either a bGreedy, an nbGreedy or an nbBuffer connector. A junction makes two successive synchronous segments independent as they can run at different iteration rates. Predicate lossy\((j)\) is true if junction \(j\) is lossy.

The next step of our automatic construction consists in the equalization of the number of junctions between all the sibling pipelines of a coherence sub-graph. This is needed to fulfill the coherence constraints. Indeed, controlling the messages entering a synchronous segment allows to control the messages at the end of the segment. To summarize, our method tends to preserve as many junctions as possible in order to preserve as many independent segments as possible from the initial graph. It also ensures that the number of independent segments is the same in all the pipelines from a source port to a destination port of the coherence. Coherence control can then be operate piecewise along them. After path segmentation, junctions of the same level will be grouped inside plateaus.

Definition 6. Let \(G\) be an application graph, \(p_1\) and \(p_2\) two sibling pipelines of a constraint \(\kappa\) in \(G\) starting at components \(C_1\) and \(C_2\) respectively and reaching component \(D\). Due to the segmentation step, \(p_1 = (S_1^1, j_1^1, \ldots, j_n^1, D)\) and \(p_2 = (S_2^1, j_1^2, \ldots, j_n^2, D)\) are composed of the same number \(n\) of synchronous segments where \(S_1^1\) (respectively \(S_2^1\)) starts at \(C_1\) (respectively \(C_2\)) and are separated by \(n - 1\) junctions \((j_i^1)_{1 \leq i \leq n-1}\) for \(p_1\) and \((j_i^2)_{1 \leq i \leq n-1}\) for \(p_2\). We say that the junctions \(j_1^1\) and \(j_1^2\) are of the same level, which is denoted \(j_1^1 \leftrightarrow j_1^2\). The reflexive-transitive closure of \(\leftrightarrow\) is denoted \(\leftrightarrow^*\). A plateau is the set of the junctions of the same equivalence class of \(\leftrightarrow^*\).

A plateau is the entry point of several synchronous segments involved in the same constraint—or in interdependent constraints. They are the points where messages circulating in different pipelines will be controlled by regulators and by input or output synchronizations as explained further in Section 3.5. Further in the process, a plateau will either:

1. be replaced by the primary regulator of the coherence, the role of which is to establish the coherence as expressed in the formulae of the constraint,
2. or be a synchronization point, maintaining the coherence of the pipelines thanks to input and output synchronization mechanisms.

The equalization of the number of junctions—and thus, of synchronous segments—between multiple pipelines is obtained by allowing the system to switch some connectors from \{sFIFO or bBuffer\} to nbBuffer, or from \{bGreedy or nbGreedy\} to \{sFIFO, bBuffer or nbBuffer\}. It is allowed, for the sake of coherence, to relax blocking, non-blocking or lossy constraints of the connection specification. However, non-lossy constraints are never relaxed. In addition, no blocking connectors can be put before a free component either. The path segmentation is solved on the whole application graph. We use a linear system where each variable is associated to a connector. The domain of the variables is \(\{0, 1\}\). 0 means that the connector is either a sFIFO or a bBuffer, and 1 any of
the three other patterns -and a potential regulator location. Since these three other patterns define junctions, it is sufficient to impose that the sums of the variables of each sibling pipeline be equal to ensure that they have the same number of segments. Additional constraints are also added to the problem to avoid misleading solutions. For each connector \( c \) of \( G \), according to the properties of the corresponding connection in the specification graph and those of the sender and the receiver, we determine the set of compatible patterns. If this set contains only elements of \{nbBuffer, bGreedy, nbGreedy\}, we add \( v_c = 1 \) to the linear system \( Eq_G \).

In this process, it is also crucial to anticipate the placing of the regulators as they will replace plateaus. One regulator is sufficient for a coherence constraint and it will be crossed by all the sibling pipelines so that it can compare their message iterations and adjust their flows. This regulator is called the primary regulator of the coherence in contrast with other regulators the pipelines might come across and that may be set to control another coherence.

**Definition 7.** Let \( \kappa \) be a coherence constraint, \( g_\kappa = \{p_1 \ldots p_z\} \) its subgraph and \( \Pi = \{J^1 \ldots J^n\} \) the set of plateaus of \( g_\kappa \) such that \( J^i = \{j^i_1 \ldots j^i_z\} \). \( i \in [1,n] \) denotes the level of the plateau \( J^i \) along the pipelines of \( g_\kappa \). \( J^i \) is a location candidate for the primary regulator of \( \kappa \) if \( \exists j^i_k \in J^i \) and \( \exists p_k \in g_\kappa \) such that \( \text{lossy}(j^i_k) = \text{true} \) and \( j^i_k \in p_k \) and \( \forall p_l \in \text{siblings}(p_k), \exists C \in p_l \cap p_k \) such that \( C \) is a component and \( \text{rank}_{p_l}(C) < \text{rank}_{p_k}(j^i_k) \). Then, \( J^i \in \Pi \) is the primary regulator location for \( \kappa \) if \( \exists J^j \in \Pi \) a primary regulator candidate for \( \kappa \) such that \( j < i \).

![Fig. 5. Simple illustration of the regulator setup policy](image)

The definition of the regulator given in Section 2.3 requires the junctions the primary regulator replaces to be lossy. Consequently, the highest junctions in a
coherence subgraph before setting the regulator have to be lossy. In addition, to
respect a coherence constraint, data must not be lost before the primary regu-
lator. Otherwise \( it(m_1) \neq it(ori(m_1)) \) for a message \( m_1 \) reaching the regulator
and it would not be possible to express a constraint on \( it(ori(m_1)) \) in the
primary regulator anymore. Thus, there must not be other junctions above the first
lossy ones on the pipelines. For that, the system forces all the connectors pre-
ceding the highest lossy junctions to form a synchronous segment by enforcing
\( v_c = 0 \) for each of them. The primary regulator has also to be set before any
intersection between two sibling pipelines. Otherwise, the iteration number of
the messages produced by the common component would not allow to distinct
the message iterations from the two sources of the sibling pipelines anymore.

Figure 5 shows a sample application in which we consider coherence between
the two input ports of component \( G \) is requested with respect to the outputs
of \( A \) and \( B \). The regulator has three possible locations represented by plateaus
1, 2 and 3. Obviously, plateau 3 is not convenient as part of the flows from \( A \)
and \( B \) merge at \( F \) and become indistinguishable. To guarantee performance, the
primary regulator has also to be set as close as possible to the sources of the
involved pipelines in order to release the synchronicity as soon as possible. For
example, if the primary regulator is set at plateau 2 in Figure 5, the junctions
of plateau 1 will necessarily be removed and replaced by synchronous connections.
Consequently, the primary regulator will rather be set by the system at plateau
1 so that the desynchronization plateau 2 can be kept.

At this step, if a pipeline appears not to have any lossy connector at all, it
will prevent the establishment of the provenance coherence. A warning that a
tight coherence [10] can be ensured instead is then raised. The set of additional
equations in the linear system is denoted \( Fix_G \). Most of the time, the system
has many solutions that are not equivalent from a performance point of view.
We then give priority to those that maximize the application’s performance, i.e.
that preserve at best the initial junctions. This is expressed by the objective
function \( \text{Maximize}(\text{Sum}(J_G)) \) where \( J_G \) is the set of junctions initially set in
\( G \) and \( \text{Sum}(J_G) = \sum_{c \in J_G} (v_c) \). So the linear problem we want to solve is \( Eq_G \cup
Fix_G \cup \text{Maximize}(\text{Sum}(J_G)) \).

After the numbers of junctions in the pipelines were made the same, it be-
comes possible to definitively set the type of each junction. First, plateaus are
formed according to Definition 6. Plateaus belonging to different coherences are
grouped if they have at least one connector in common. Then, as demonstrated
in [10], the connectors of a given plateau must be of the same type to effec-
tively maintain the coherences all the way down to the destination input ports.
When a plateau contains connectors of different types, we set all its connectors to
nbBuffer if it contains at least one nbBuffer pattern and, otherwise, to nbGreedy
if it contains at least one connector of this type.

3.4 Regulator setup

This step sets the necessary regulators to cover all the coherence constraints. The
system iterates over the provenance coherence constraints, setting their primary
regulators one by one. If the selected plateau is of type nbGreedy, the regulator will adopt a non-blocking policy on all its output ports, and that for a matter of coherence between them. Otherwise, it will be blocking on all its output ports.

The filtering rules inscribed inside a regulator are adapted to the location of the regulator along the pipelines. Therefore, for each input port of the regulator, the source output port of the pipeline is sought and a rule with respect to sibling pipelines is added. Here, because of merged plateaus, a regulator of a coherence may intersect pipelines of other coherences but without being their primary regulator. It then automatically adds equality rules between all the pipelines which are siblings with respect to other coherences in order to maintain them. In Figure 7, not only does regulator $R_2$ ensure $\kappa_1$ but it also maintains $\kappa_2$ established by $R_1$. More formally, let $r$ be the primary regulator of a coherence constraint $\kappa = D_{i_1,i_2} : C_{1.o_1} \circ \alpha \times C_{2.o_2} + \delta$. $F(r)$ consists in the set of filtering rules $f_{p_k,p_l}$ where $p_k \subset P_k$ and $p_l \subset P_l$ such that $P_k$ and $P_l$ are two sibling pipelines and $p_k, p_l$ reach respectively ports $in_k$ and $in_l$ of $r$. $f_{p_k,p_l} = in_k \circ f \alpha f \times in_l + \delta f$, where $\circ f = \circ, \alpha f = \alpha$ and $\delta f = \delta$ if $P_k$ and $P_l$ are sibling with respect to $\kappa$ and $\circ f = \circ = \cdot$, $\alpha f = 1, \delta f = 0$ otherwise.

3.5 Coherence preservation

The coherence between sibling pipelines established by the regulators has to be maintained until the final input ports. This is achieved by setting up, in the remaining plateaus, the tight coherence mechanisms introduced in [10].

**Definition 8.** We denote by $M$ a series of messages, by $|M|$ its length and $m^i$ denotes its $i^{th}$ message. A set of series of messages $\{M_1, \ldots, M_n\}$ is called synchronized if $|M_1| = \cdots = |M_n|$ and $\forall i \in [1,|M_1|], it(m^i_1) = \cdots = it(m^n_i)$.

The synchronicity mechanisms consist in input and output synchronization patterns. While the output synchronization mechanism remains as defined in [10], we slightly enrich the input synchronization pattern so that it can also begin with a regulator instead of two junctions.

**Definition 9.** In an application graph, an input synchronization is a composition pattern that consists of two synchronous segments $p_1, p_2$ of respectively $k$ and $l$ components and ended by respectively the components $C_{1.k}$ and $C_{2.l}$ not necessarily distinct and

- either two junctions $j_1, j_2$ of the same type and not necessarily distinct, triggered by their receivers $C_{1.k}$ and $C_{2.l}$ and a backward cross-triggering consisting of $(C_{1.k}.e,j_2.s)$ and $(C_{2.l}.e,j_1.s)$.
- or a regulator $r$ triggered by $C_{1.k}$ and $C_{2.l}$ and a backward cross-triggering consisting of $(C_{1.k}.e,r.s)$ and $(C_{2.k}.e,r.s)$.

This pattern is denoted $J_*(p_1,p_2)$. 222
The input synchronization ensures that the junctions \( j_1, j_2 \) belonging to a plateau \( P \) of junctions select their messages at the same time and that no new messages are accepted by the first components before all the components of the segments are ready for a new iteration. If \( P \) is a regulator, it may alter the message flows such that the messages entering \( p_1 \) and \( p_2 \) are coherent with respect to the rules inscribed in it. The simultaneous triggering preserves the synchronicity of the pipelines and of the dispatched messages. If \( P \) is non-blocking and does not contain a pair of messages for \( p_1 \) and \( p_2 \) when it is triggered, it issues a couple of empty messages instead. Figure 6 shows the different input synchronization cases that can be met according to the degree of merging of \( p_1 \) and \( p_2 \). In Figure 6(a), \( p_1 \) and \( p_2 \) begin with the same component so only two triggering links are needed. In Figure 6(b), \( p_1 \) and \( p_2 \) have two distinct sources. In case there is a regulator instead of the junctions as in Figure 6(c), it is triggered by the components that are its direct receivers. In Figure 6(d), the pipelines are merged before they reach the junction. Their synchronization is then implicit.

**Fig. 6.** There are five different input synchronization cases

**Definition 10.** In an application graph, an output synchronization is a composition pattern involving

- two synchronous segments \( p_1 \) and \( p_2 \) not necessarily distinct of respectively \( k \) and \( l \) components and ended by respectively components \( C^k_1 \) and \( C^l_2 \),
- two bBuffer connectors \( bB_1 \) and \( bB_2 \) following respectively \( p_1 \) and \( p_2 \),
- a forward cross-triggering consisting of \((C^k_1.e,bB_2.s)\) and \((C^l_2.e,bB_1.s)\).

This pattern is denoted \((p_1,p_2) \ast bB\).

This composition pattern ensures that the delay between the synchronous segments to produce messages is absorbed. As the bBuffer connectors select their messages at the same time when all the last components of the synchronous segments are done, the messages are also delivered at the same time. Note that this property is maintained when the two bBuffer connectors are triggered by a same additional set of signals. If \( C^k_1 = C^l_2 \), no additional bBuffers or forward cross-triggering is needed as \( p_1 \) and \( p_2 \) are naturally synchronized by this common destination component. Moreover, no output synchronization is needed if \( p_1 \) and \( p_2 \) precede a regulator as the regulator itself buffers incoming messages and outputs and guarantees the simultaneity of these outputs.

In what follows we demonstrate that the different steps of our construction generate an application graph which respects the coherence constraints.
Definition 11. In an application graph $G$, the composition $J^* (s_1, s_2) * bB$ where $s_1$ and $s_2$ are two synchronous segments is called a pair of coherent segments. $[J^* (s_1, s_2) * bB]^q$ denotes the composition of $q$ coherent segments $J^1 * (s_1^1, s_2^1) * bB^1 * \cdots * J^q * (s_1^q, s_2^q) * bB^q$.

Theorem 1. Let $G$ be an application graph and $(S_1, S_2) = [J^* (s_1, s_2) * bB]^q$ two segments in $G$. If the series of messages $M_1$ and $M_2$ stored in the junctions $j_1^1$ and $j_2^1$ of the first coherent segments are synchronized, then the set of messages $m_1$ and $m_2$ stored respectively in the bBuffer connectors $bB_1^1$ and $bB_2^1$ of the last coherent segments are such that $it(m_1) = it(m_2)$ and $it(ori_{S_1}(m_1)) = it(ori_{S_2}(m_2))$ when the bBuffers are triggered.

This theorem comes from [10] where no regulators existed. This result can be easily extended to the case where some junctions ($j_1^1, j_2^1$) ($1 \leq k \leq q$) are replaced by a regulator with two input ports $in_1^k$ and $in_2^k$ and that imposes $m_1^k = in_2^k$. Such a constraint plays, indeed, the same role as an input synchronization.

Theorem 2. Let $P_1$ and $P_2$ be two sibling pipelines of a coherence constraint $\kappa = D_{i_1, i_2 : C_1, \alpha} \circ \alpha \times C_2, \delta$. Let $m_1$ and $m_2$ two messages read by $D$ at the same iteration on respectively $i_1$ and $i_2$ ports. Then $m_1$ and $m_2$ verify that $it(ori_{P_1}(m_1)) = it(ori_{P_2}(m_2)) + \delta$.

Proof. Since $P_1$ and $P_2$ are two sibling pipelines of the constraint $\kappa$, we can decompose them into $n + 1$ pairs of coherent segments as follows: $(P_1, P_2) = (p_1, p_2) * J * (C_1', P_1', C_2' P_2')$ with

- $(p_1, p_2)$ two synchronous segments such that the first edge of $p_1$ is connected to $C_1, \alpha$, and the first edge of $p_2$ is connected to $C_2, \alpha$,
- $J$ is in the plateau that is the primary regulator $r$ of $\kappa$ where $p_1$ is connected to port $i_1$ and $p_2$ to port $i_2$ of $r$,
- $C_1', P_1'$ and $C_2', P_2'$ are composed of synchronous segments, begin with components $C_1'$ and $C_2'$ respectively and end at respectively $i_1$ and $i_2$ of $D$.

Let $m_1$ and $m_2$ be two messages read at the same iteration of $D$ on respectively ports $i_1$ and $i_2$. Since $J$ is the primary regulator of $\kappa$, all the other regulators crossed by $P_1'$ and $P_2'$ impose equality on the ports that concerns $P_1'$ and $P_2'$. Therefore, from Theorem 1, $it(ori_{P_1'}(m_1)) = it(ori_{P_2'}(m_2))$.

Since $r$ is the primary regulator of $\kappa$, the rule $f_{P_1, P_2} = in_1 \circ \alpha \times in_2 + \delta$ is in $F(r)$. Therefore, the messages $ori_{P_1'}(m_1)$ and $ori_{P_2'}(m_2)$ belong to a set of messages that validates $F(r)$. This means that we have $it(ori_{C_1', P_1'}(m_1)) = \alpha \times it(ori_{C_2', P_2'}(m_2)) + \delta$.

Since $p_1$ and $p_2$ are two synchronous segments, we know that for any messages $m_1'$ and $m_2'$ reaching ports $in_1$ and $in_2$ of $J$, we have $it(ori_{P_1}(m_1')) = it(m_1')$ and $it(ori_{P_2}(m_2')) = it(m_2')$. From that, we can conclude that $it(ori_{P_1}(m_1)) = \alpha \times it(ori_{P_2}(m_2)) + \delta$.

This theorem proves that the coherence $\kappa$ is respected in the application graph automatically constructed.
Figure 7 gives the final application graph under two coherence constraints $\kappa_1$ and $\kappa_2$ of the application specified in Figure 4. To put coherence preservation in practice, the system first adds the backward cross-triggerings to the junctions. Since a plateau may involve more than two segments, our construction generalizes Definition 9. For a plateau $j_1 \ldots j_n$ and the segments $p_1 \ldots p_n$ ending with components $C_1 \ldots C_n$, we add the set of edges $\{(C_i.e,J_j.s)\mid i \neq j\}$. On the example, regulator $R_1$ is synchronized by $D$ and $E$ and regulator $R_2$ is synchronized by $G$, $H$ and $I$. Then, to implement the output synchronization, we add one bBuffer connector just after each $C_i$ ($i \in [1,n]$) and add the edges for the forward cross-triggerings. Output synchronization mechanisms can be noticed before plateaus $\{c_9,c_{10}\}$ and $\{c_{11},c_{12}\}$. These plateaus are also subjected to input synchronization from respectively components $J$ and $K$.

4 Discussion and future work

The great emphasis on performance in the communication between the components of our model targets the building of real-time interactive scientific visualization applications, a particular type of scientific applications to which, to
our knowledge, no specific component model is dedicated yet. The intended interactivity in these applications is not limited to a passive manipulation of the graphical output. It is rather active and its effects are propagated throughout the whole running application.

In our approach, we associated to the commonly known spatial provenance its temporal dimension and used it to ensure coherence in a loosely connected system. Provenance-based coherence expands the definition of the tight coherence introduced in [10] allowing the specifications of finer rules. We presented a method to automatically set regulators and connectors to fulfill coherence constraints. As explained, our method sets the regulators as high as possible in the graph to allow the greatest number of desynchronized segments in the pipelines and thus, promote performance. This can however cause one regulator to be primary for multiple coherence constraints having the same sources. The potential drawback of this situation is having conflicting coherence rules inside the same regulator. Fortunately, unless the conflicting rules are equality+offset rules, the regulator will always output messages, reflecting their lowest common denominator. It is also assumed that a buffer, whether inside a connector or a regulator, has an infinite capacity. At implementation, we are considering the use of a performance model to obtain runtime adaptive buffer sizes. This would, in addition, make non-lossy channels possible in regulators. To prevent overflows, it can also be noticed that our construction method sets, as a priority, sFIFOs and Greedies before Buffers.

The implementation of the complete application generator is ongoing. Meanwhile, representing components, connectors, regulators and small applications as Petri nets [19] serves as a temporary and light model checking means. Our objective is to provide a SWMS specifically designed for real-time interactivity. The current paper addresses the application composition phase and not component programming. Solutions for the latter, focused on code reuse, were presented in [9]. The main solution consists in a high level API to transform C, C++ or Fortran code into FlowVR [8] iterative components. FlowVR is a middleware to develop and run high-performance interactive applications.

In the future, we plan to extend our coherence constraints to data properties other than the iteration number. For example, a module may use the results of two different simulations that generate messages at different rates but stamped with simulation time. In this case the user may impose constraints on these time stamps to get coherent results. Another extension consists in expanding our component definition to supporting not only, regular message streaming but also event-based message emission.

References


Learning from Failures: a Lightweight Approach to Run-Time Behavioural Adaptation *

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Abstract. Software integration needs to face signature and behaviour incompatibilities that unavoidably arise when composing services developed by different parties. While many of such incompatibilities can be solved by applying existing software adaptation techniques, these are computationally expensive and require to know beforehand the behaviour of the services to be integrated. In this paper, we present a lightweight approach to dynamic service adaptation which does not require any previous knowledge on the behaviour of the services to be integrated. The approach itself is adaptive in the sense that an initial (possibly the most liberal) adaptor behaviour is progressively refined by learning from failures that possibly occur during service interaction.

1 Introduction

The wide adoption of Web service standards has considerably contributed to simplifying the integration of heterogeneous applications both within and across enterprise boundaries. The languages to describe messaging (SOAP), functionalities (WSDL) and orchestration of services (WS-BPEL) have been standardised, but the actual signatures and interaction protocols of services have not. For this very reason, service adaptation [2,13,17] remains one of the core issues for application integration in a variety of situations: the need of overcoming various types of mismatches among services developed by different parties; customising existing services to different types of clients; adapting legacy systems to meet new business demands; or ensuring backward compatibility of new service versions.

Various approaches have been proposed to adapt service signatures [6], process behaviour [3], quality of service [7], security [12] or service level agreements [15]. In this paper, we focus on signature and behaviour incompatibilities, whose occurrence can impede the very interoperability of services. Many signature and behaviour incompatibilities can be solved by applying existing (semi-)automated

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adaptation techniques. However such techniques present two limitations: i) they require signature and behaviour of both parties to be known before service interaction starts, and ii) they are computationally expensive since they explore the whole interaction space in order to devise adaptors capable of solving any possible behaviour mismatch.

In this paper we focus on the problem of dynamic adaptation in applications running on limited capacity devices, as in typical pervasive computing scenarios where (unanticipated) connections and disconnections of peers continuously occur. Unfortunately, the limited computing, storage, and energy resources of such devices inhibit the applicability of most existing adaptation approaches.

We present a lightweight adaptive approach to the adaptation of services that is capable of overcoming signature and behaviour mismatches that would otherwise impede service interoperation. The approach is lightweight in the sense that it requires low computing and storage capabilities to run.

The adaptation is controlled by an adaptation contract which specifies what is considered a meaningful and successful trace in a declarative manner. It also states how to solve signature incompatibilities between the known operations of the services, therefore the operations of the services (their signature) must be known but the sequence on which those operations are offered and requested (i.e., the service behaviour, as those represented by BPEL or BPMN processes) do not need to be known and might even change during the lifespan of the adaptor.

The adaptation process is itself adaptive in the sense that an initial (possibly the most liberal) adaptor behaviour is progressively refined at run-time by learning the behaviour of the services from failures that may occur during service interactions. Roughly speaking, the adaptor initially allows all interactions that satisfy the current adaptation contract. If an interaction session between the services fails w.r.t. the contract, the adaptor memorises the interaction trace that led to the failure and it will suitably exploit it to inhibit that failing trace in following sessions. Intuitively speaking, the more failures will be experimented, the faster the adaptor will refine its behaviour so as to allow only deadlock-free interactions among the services.

Learning and inhibiting erroneous traces tackle permanent failures. For instance, a behavioural incompatibility which leads the system to a deadlock situation or a hardware malfunction (maybe due to low battery) which disables part of the functionality. In addition, communications in pervasive computing can be unstable due to changes in the environment. For instance, shadow fading [10], where messages might be lost due to the presence of possibly moving obstacles, has deep impact in the reliability of communication channels. We propose several learning policies which tackle this scenario of sporadic errors. Inhibited traces learned by the adaptor are eventually forgotten so that it can re-adapt itself to drastic changes in service functionality, temporal changes in the environment or sporadic communication failures.

As one may expect, the results of the refinement performed by this adaptive adaptation approach are particularly interesting when the process starts with a non-empty adaptation contract. However, the approach can overcome message
ordering mismatches [11] also in the extreme situation in which no such adaptation contract is available. When compared with the few other existing proposals of lightweight behaviour adaptation of services, such as [5] for instance, our approach features the important advantage of requiring just an adaptation contract based on the services signatures, it does not require to know the interaction behaviour of the services that need adaptation. In other words, the adaptor is not synthesised at design time, instead, it is directly deployed with no other information than an adaptation contract and it will successively learn the behaviour of the services and how to solve their behavioural incompatibilities.

Failed traces, although undesirable, are commonplace in pervasive computing due to its unstable nature, and unavoidable due to communication errors. Therefore, services in these unstable scenarios are usually able to autonomously restart whenever failures happen so, in general, it is safe to assume some failed traces. As regards the complexity in time and space of learning adaptors, these only depend on the size and structure of the adaptation contract.

The structure of the paper is the following. We introduce behavioural adaptation in Sect. 2. The lightweight adaptive approach to dynamic service adaptation is formally presented in Sect. 3 and we comment on several learning policies in Sect. 4. Then we proceed to evaluate its implementation with an example based on two real-world data-diffusion protocols for sensor networks (Sect. 5). Some related work is introduced in Sect. 6 and we finally conclude with Sect. 7.

2 Behavioural Adaptation

The deployment of suitable “adapters-in-the-middle” has proven to be an effective way to overcome signature and behaviour incompatibilities between services [3]. Intuitively speaking, such adaptors intercept, collect, and modify the messages exchanged by two parties so as to overcome their incompatibilities. The adaptor behaviour is specified by an adaptation contract defining a set of correspondence rules between actions and (optionally) some constraints on the use of such rules.

Definition 1. An adaptation contract $C$ is a finite state machine (FSM, for short) $(\Sigma^c, S^c, s_0^c, F^c, T^c)$ where $\Sigma^c$ is a set of correspondence rules, $S^c$ is a set of states, $s_0^c \in S^c$ is the initial state, $F^c \subseteq S^c$ is the set of final states, and $T^c \subseteq (S^c \times \Sigma^c \times S^c)$ is a set of labelled transitions. Correspondence rules in $\Sigma^c$ have the form $a \diamond b$ where: $a$ and $b$ are input or output communication actions; one side of the rule can be empty (viz., $a \diamond$ or $\diamond b$); and if both $a$ and $b$ are present, then one is an input action and the other is an output action.

Adapters act as mediators between two sides. Any communication between those sides must be intercepted and handled by the adaptor. Actions on the each side of correspondence rules denote the complementary actions that the adaptor will perform towards the service on that side. For instance, a correspondence rule such as $!msg \diamond ?msg'$ (where $msg$ and $msg'$ are operation names followed by symbolic parameters) states that if the adaptor receives message $msg$ from the service on the left-hand side then it will have to (eventually) send message $msg'$

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to the service on the right-hand side. Every message received by the adaptor is matched against a correspondence rule, and such matching possibly updates the state of stored parameters maintained by the adaptor. Once correspondence rule ![msg ?msg] is triggered, message msg is instantiated and inserted in a queue of messages to be eventually sent. When the target service will be ready to receive, the first matching message in the queue can be delivered.

The transition relation $T^c$ imposes further (optional) restrictions on the order in which correspondence rules can be triggered. In this way, $T^c$ permits to enforce high level policies on the communication such as “do not perform more than three requests” or “after every request there must be an acknowledgment”.

**Example 1.** Our running example is based on a simplified meteorologic system. We have three incompatible services but with complementary functionality: a) a temperature sensor service, this service could be deployed in a sink of a temperature sensor network; b) a monitoring service which registers the information, this could be located in a laptop; and c) a humidity service which might be deployed in the same infrastructure as the temperature sensor network or otherwise.

The signatures of the services (i.e., their operation names and arguments) are known. The temperature service (service a) has output operations ![user(usr)] and ![pass(psw)] to authenticate with its user name (argument usr) and password (psw); an operation to notify of the current temperature, i.e., ![upload(temp)]; two input operations for the upload to be either ![denied()] or answered with a new interval of time prior the next notification (![delay(time)]); and finally, an output operation to notify that it finishes its current session, ![end()]. Intuitively speaking, input actions (e.g., ![denied()]) represent the availability of service operations while output actions represent service requests (e.g., ![upload(temp)]), both with the types of their arguments between parentheses.

The monitoring service (service b) might be a new version or come from a different vendor so that it has operations with similar functionalities but incompatible signature. Instead of operations ![user(usr)] and ![pass(psw)] expected by service a, it has a single authentication operation ![login(usr, psw)]. The authentication can be ![rejected()] or ![connected()]. It receives the temperature notifications with an operation ![register(temp)] and it sends the answer always through ![answer(time)]. This service can receive a ![quit()] petition and it notifies of the finished session with ![end()]. The monitoring service requires humidity information (typed humid) before deciding how long to wait for the next temperature update. For this reason, it requests the humidity information to the humidity service (service c) through the request and response ![getHumid()] and ![getHumid(humid)]. The latter is understood by service c but, instead of the former, service c needs the temperature information to do some calibration via ![getHumid(temp)] and it finally ends its session with ![finish()].

Figure 1 illustrates a possible adaptation contract for these services. Rule $v_a$ enables the adaptor to receive action user and refers to its argument as $U$. Rule $v_1$ first receives the password (in P) with action pass and, as a consequence, it eventually sends a login message with both the user $U$ and password $P$ previously received. The rest of the correspondence rules behave accordingly. The
\( \Sigma^c = \{ !\text{user}(U) \implies (v_u), \quad !\text{pass}(P) \implies ?\text{login}(U, P) \implies (v_l), \quad \} \\
\quad \implies !\text{connected()} \implies (v_c), \quad !\text{upload}(D) \implies (v_p), \quad \} \\
\quad \implies ?\text{register}(D) \implies (v_r), \quad ?\text{getHumid}(D) \implies !\text{getHumid()} \implies (v_h), \quad \} \\
\quad \implies ?\text{delay}(T) \implies !\text{answer}(T) \implies (v_d), \quad !\text{getHumid}(H) \implies ?\text{getHumid}(H) \implies (v_g), \quad \} \\
\quad \implies ?\text{denied()} \implies !\text{rejected()} \implies (v_d), \quad \implies !\text{finish()} \implies (v_f), \quad \} \)

(a) Correspondence rules

\( \Sigma^c \setminus \{v_u, v_c, v_g, v_f\} \quad \Sigma^c \setminus \{v_r, v_d, v_e\} \)

(b) Contract FSM

Fig. 1. An adaptation contract.

As we have seen in the example, services can employ different alphabets of actions (different names of actions as well as different names, number or order of parameters). The synchronisation rules of the contract (\( \Sigma^c \)) tell us how to solve these signature incompatibilities. In addition, services might also lock due to behavioural incompatibilities between them. These incompatibilities arise because one service offers and requests operations in a different order than the one expected by another.

The intentional semantics of the contract specifies what we want from the adapted system (i.e., what are the desired interactions between the services) but it does not necessarily say anything about how to solve behavioural incompatibilities (since the behaviour of the services might be unknown). In order to adapt behavioural incompatibilities without knowing the actual behaviour of the services (which might even change drastically due hardware problems or low battery, for instance) the runtime adaptors presented in this paper must learn to be compliant with the given adaptation contract (i.e., to respect the intentional semantics of its contract) and avoid the deadlocks that might occur due to incompatibilities between the unknown behaviour of the services.
2.1 Intensional Semantics of Adaptation Contracts

The intensional semantics of an adaptation contract provides the interactions between the services and the adaptor allowed by the contract. Formally, the intensional semantics of an adaptation contract \( c \) is defined by a labelled transition system \( \Delta_{c} \) over configurations of the form \( \langle s, \Delta \rangle \) where \( s \) is the current state of the contract and \( \Delta \) is a multiset of pending actions that the adaptor will have to eventually perform. A transition \( \langle s, \Delta \rangle \xrightarrow{\Delta_{c}} \langle s', \Delta' \rangle \) indicates that an adaptor could, by contract \( c \), execute action \( x \) in state \( s \) with pending actions \( \Delta \). The transition system \( \Delta_{c} \) is defined by the following inference rules

\[
\begin{align*}
(1) & \quad \frac{(s, a \circ b, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \cup \{b\} \rangle} \\
(12) & \quad \frac{(s, a \circ b, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \cup \{\pi\} \rangle} \\
(13) & \quad \frac{(s, \Delta \cup \{x\}) \xrightarrow{\Delta_{c}} (s, \Delta)}{(s, \Delta \cup \{x\}) \xrightarrow{\Delta_{c}} (s, \Delta)} \\
(14) & \quad \frac{(s, a \circ c, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \rangle} \\
(15) & \quad \frac{(s, a \circ b, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \rangle}
\end{align*}
\]

where the complementary action of a non-internal action \( a \) is denoted by \( \pi \) (e.g., if \( a = \text{!do()} \) then \( \pi = \text{?do()} \), and vice-versa).

Note that the labels denoting the actions of the adaptor are annotated with a left-hand or right-hand bar to explicitly represent whether they are communication actions performed by the adaptor towards the service on the left-hand side \( (|a) \) or towards the service on the right-hand side \( (b) \), respectively. Note also that an ordered semantics of pending actions is assumed, that is, in rule (13) we assume that if there is more than one \( x \) in the multiset \( \Delta \), then the emitted \( x \) is the oldest in \( \Delta \). Finally, since in a correspondence rule \( a \circ b \) of an adaptation contract either \( a \) or \( b \) may be absent, the definition of \( \Delta_{c} \) includes also the following rules:

\[
\begin{align*}
(14) & \quad \frac{(s, a \circ c, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \rangle} \\
(15) & \quad \frac{(s, a \circ b, s') \in T^c}{\langle s, \Delta \rangle \xrightarrow{\sigma_{c}} \langle s', \Delta \rangle} \\
\end{align*}
\]

It is worth noting that the intensional semantics defined by rules (11) to (15) may force eager choices. Such eager choices may occur, among other situations, when an adaptation contract contains more than one correspondence rule for an action \( a \). Consider for instance the simple contract \( c = \{\Sigma^c, S^c, s^c_0, F^c, T^c\} \) where \( \Sigma^c = \{a \circ b, a \circ c\} \), \( S^c = \{s_0, s_1\} \), \( s^c_0 = s_0, F^c = \{s_1\} \), and \( T^c = \{(s_0, a \circ b, s_1), (s_0, a \circ c, s_1)\} \). Then, two of the transitions that may fire in the initial state, namely \( \langle s_0, \emptyset \rangle \xrightarrow{\sigma_{c}} \langle s_1, \{b\} \rangle \) and \( \langle s_0, \emptyset \rangle \xrightarrow{\sigma_{c}} \langle s_1, \{c\} \rangle \), create an eager choice of the adaptor, which must pick one of them when it executes \( a \). Intuitively, such an unnecessary eager choice may lead the adaptor to fail adapting some interactions. We could enforce contracts to be deterministic but, instead, we allow such flexibility by providing a lazy choice alternative which results in deterministic adaptors. Lazy choice is modelled by lifting transition system \( \Delta_{c} \) so as to deal with sets of pairs \( \langle s, \Delta \rangle \).

\[
\begin{align*}
(1) & \quad A' = \{\langle s', \Delta' \rangle \mid \exists (s, \Delta) \in A. \langle s, \Delta \rangle \xrightarrow{\Delta_{c}} \langle s', \Delta' \rangle \} \neq \emptyset \\
A \xrightarrow{\Delta_{c}} A' \\
\end{align*}
\]

Not every execution order among the correspondence rules in the contract avoid deadlocks since deadlocks depend on the actual behaviour of the services,
which is unknown. For instance, let us consider the actual behaviour of the services of our running example were the FSMs depicted in Fig. 2. Internal choices (e.g., if-then-else or switch conditionals) are modelled by τ actions as usual, while external choices (e.g., WS-BPEL pick) are modelled by input-action-labelled transitions leaving from the same state. Now, assuming that service b internally decides to connect (it follows the right-hand side τ), then the intensional semantics of the contract in Fig. 1 allows the sequence of rules v_u : v_s : v_p : v_q (which, among others, corresponds to trace |?u| ?s : !l| |?c| |?p| ?q where actions are represented by their underlined characters and ‘::’ is the append operator). This sequence would lead the system to a deadlock because, in that point, service b cannot participate in the rules needed for service a to reach a final state (i.e., v_d and v_s, at least). Because of these deadlock situations, the intensional semantics of adaptation contracts are refined into a concrete adaptor behaviour capable of controlling the services and leading them to successful states while avoiding locks. This refinement is the key concept of traditional adaptor synthesis proposals [1,3,11,13]. These related works, however, are focused on design time and they require to know in advance the behaviour of the services. Unlike those related works, the goal of the learning process presented in this paper is to do this adaptation at run time without knowing the behaviour of the services.

Example 2. Assuming that the unknown behaviour of the services were the FSMs shown in Fig. 2, the most general adaptor compliant with the contract in Fig. 1 would be the one depicted in Fig. 3. For the sake of clarity, actions in Fig. 3 have been reduced to their underlined letters in the contract and have been prefixed with the identification of the communicating service. Such an adaptor could be generated using traditional approaches being provided the contract and the behaviour of the services. The learning adaptors presented in this work do not need to know the behaviour of the services. A learning adaptor for our running example dynamically learns to synchronise with the services in the same way that the adaptor in Fig. 3 does. When the learning adaptor converges, each of its transitions is either one in Fig. 3 or it is offered but never used by the services. In addition, the behaviour of the adaptor does not need to be stored since every transition is generated on-demand.

3 Learning Adapters

Our proposal is to directly deploy an adaptor without no other information than the adaptation contract and then the adaptor will dynamically learn the behaviour and incompatibilities of the services. The approach is to initially support every communication allowed by the adaptation contract without any guarantee about the successful termination of the current session. The adaptor learns which sessions ended correctly and, on failures, it will forbid the last communication which led to the failure. The goal is to make this process converge to the most general adaptor which complies with the adaptation contract and the given services. However, depending on the contract and the services (they
Fig. 2. The (unknown) behaviour of the services of our running example

might not be controllable due to their internal choices) it is possible that no such
an adaptor exists. In this case, the process will converge to an empty adaptor
(single initial state with no transitions) where no communication is allowed.

The following transition system $\xrightarrow{\delta}$ models the way in which an adaptor
wraps the service it adapts and interacts with the rest of the environment. An
adaptor wrapping a service according to an adaptation contract $c$ is denoted in
the transition system $\xrightarrow{\delta}$ by a term of the form: $\langle A, I, t \rangle_c [P]$ where $A$ is a set of
pairs $(s, \Delta)$ ($s$ is a state of the contract and $\Delta$ the multiset of pending actions
that it should eventually perform), $I$ is a sequence of inhibited traces that have
previously led to unsuccessful interactions according to what the adaptor has
learned so far, $t$ is the trace of actions executed so far by the adaptor during the
current interaction session, $c$ is the adaptation contract and $P$ is the current state
of the service being adapted (which is not known by the adaptor). An adaptor
at the beginning of a session is denoted by $\langle A_0, I, \lambda \rangle_c [P]$ where $A_0 = \{ (s_0, \emptyset) \}$
and $\lambda$ is the empty trace. If the adaptor has not learned anything yet, then $I$ is
empty.

$I$ contains elements of the domain of inhibited traces. Each of these trace
is a sequence of communication actions ranging over $\Sigma^*$ (where $\Sigma$ represents

(a) Service a, temperature sensor
(b) Service b, monitoring station
(c) Service c, humidity sensor
Fig. 3. Static most-general adaptor compliant with the contract and services shown in Fig. 1 and Fig. 2, respectively.

a global set of communicating actions). Among others, $I$ can be modelled as a set, a sequence or a tree. Independently of its implementation, we will denote as $I \vdash t$ whether trace $t$ is inhibited by $I$ or not. When no trace is inhibited we say that $I = \emptyset$ and we say that $I \cup I' \vdash t$ if $I \vdash t$ or $I' \vdash t$.

We will denote by $t : a$ the sequence obtained by appending element $a$ to sequence $t$, by $a \triangleleft t$ the sequence obtained by prefixing element $a$ to sequence $t$, and by $t :: t'$ the sequence obtained by concatenating sequences $t$ and $t'$. We will also say that sequence $t$ is a prefix of $t :: t'$, where both $t$ and $t'$ can be empty, being $\lambda$ the empty sequence.

Rules (Ext) and (Int) describe the steps that the adaptor can make by offering a communication to the external environment and by interacting with the service it wraps, respectively.

\[
\begin{align*}
\text{(Ext)} & \quad A \xymatrix@1{\ar[r]^{a} & A'} \land I \nmid t : [a] \\
\text{(Int)} & \quad A \xymatrix@1{\ar[r]^{b} & A'} \land P \xymatrix@1{\ar[r]^{[a]} & P'} \land I \nmid t : [b]
\end{align*}
\]

Note that the communications offered by the adaptor only depend on the current state of the adaptor, not the services. Rule (Int) models synchronisations between the adaptor and the service to be adapted as silent actions $\tau$ as such interactions are not visible by the external environment. Also the internal steps independently made by the wrapped service are modelled as silent actions (Tau). Rules (Syn) and (Par) model (commutative) parallel composition between services and adaptors with synchronous communications in the standard way:

\[
\begin{align*}
\text{(Tau)} & \quad P \xymatrix@1{\ar[r]^{[a]} & P'} \\
\text{(Syn)} & \quad P \xymatrix@1{\ar[r]^{[a]} & P'} \land Q \xymatrix@1{\ar[r]^{[b]} & Q'} \\
\text{(Par)} & \quad P \xymatrix@1{\ar[r]^{[a]} & P'} \land Q \xymatrix@1{\ar[r]^{[b]} & Q'} \\
\end{align*}
\]

By rule (Ok), an adaptor can consider an interaction session successfully terminated when it is in a final state of the adaptation contract and there are no more pending communications to perform. Let $OK_c = \{ \langle s, \emptyset \rangle \mid s \in F^c \}$. 

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A ∩ OK_c ≠ ∅ ∧ A_0 = {⟨s_{0}', ∅⟩}

(OK) (A, I, t)_{c}^{P} \xrightarrow{ok(t)} [A_0, I, \lambda]_{c}^{P}

Rule (LEARN) describes how an adaptor can autonomously decide, after a timed wait, to inhibit the trace corresponding to an interaction session that has not (yet) successfully terminated.

(LEARN) A ∩ OK_c = ∅ ∧ A_0 = {⟨s_{0}', ∅⟩}

(LEARN) (A, I, t)_{c}^{P} \xrightarrow{add(t, I)} [A_0, add(t, I), \lambda]_{c}^{P}

Note that rule (LEARN) does not constrain the way in which timed waits will be actually realised in the underlying implementation. From the viewpoint of the external environment, a learning step made by the adaptor is an internal action of the latter which may take place at virtually any moment. In Sect. 4 we will show different definitions of add(t, I) that can be employed to define different learning policies for rule (LEARN). Function add consists of including the new trace into I so that add(t, I) ⊢ t. For instance, we can define an add_0 such that:

add_0(t, I) ⊢ t' iff I ⊢ t' ∨ t' = t

Note also that rules (OK) and (LEARN) specify that the adaptor will be restarted (to its initial state A_0) when it detects the successful termination of an interaction session or when it performs a learning step^3.

A natural assumption on the services deployed in limited capacity devices is that they their behaviour is bounded in length. This does not necessarily mean that the services will expire but, instead, it means that the interaction with the service are divided in finite sessions that can be run over and over again. In the sequel we always assume bounded services whose behaviour consists of a finite set of finite length traces.

Informally, we say that a learning function add is monotonic if add(t, I) inhibits (when used in rules (EXT) and (INT)) all the traces inhibited by I. Of course, add mapping must also inhibit trace t. In order to formalize this monotonicity notion, we need to introduce the set of traces prefixed by elements of I as follows:

prefixed(I) = \{u ∈ \Sigma^* | ∃t . I ⊢ t and t prefix of u\}

Definition 2. A learning function add is monotonic if add(t, I) is a monotonic extension of I and t ∈ prefixed(add(t, I)), for each t and I. We say that add(t, I) is a monotonic extension of I (I ⊑ add(t, I)) if prefixed(I) ⊆ prefixed(add(t, I)).

^3 Rules (OK) and (LEARN) do not enforce an immediate restart of the wrapped service P and of the service Q interacting with P through the adapter in a configuration Q||[A, I, t]_{c}^{P}. We assume that the restart of P and Q is autonomously performed (by a timeout, for instance) alternatively, it can be triggered by the adaptation contract itself, which can include explicit restart messages.
We say that $\text{add}(t, I)$ is a proper monotonic extension of $I$ ($I \sqsubseteq \text{add}(t, I)$) if $\text{prefixed}(I) \sqsubseteq \text{prefixed}(\text{add}(t, I))$.

Obviously, $\sqsubseteq$ relationship defined on sequences of traces is a pre-order

We now prove that the adaptation process converges if a monotonic learning function $\text{add}$ is employed in rule (LEARN) to adapt bounded services.

**Proposition 1 (Convergence).** Let $S$ and $P$ be two bounded services, $A_0$ be an adaptor for contract $c$ in its initial state $A_0 = \{\langle s_0, \emptyset \rangle \}$, and $I_0$ be a (possibly empty) domain of inhibited traces. If the adaptor employs a monotonic learning function, then there exists a sequence $I_0, I_1, \ldots, I_n$, with a finite $n \geq 0$, such that:

1. $\forall j \in [0, n) \quad \exists S', P'$. with $I_j \subseteq I_{j+1}$, and $S|\langle A_0, I_j, \lambda \rangle_c[P] \xrightarrow{\tau}^* \langle A_0, I_{j+1}, \lambda \rangle_c[P']$

2. $\exists S', P', I_{n+1}$. with $I_n \sqsubset I_{n+1}$.

The previous proposition shows that the training process with bounded services is finite and it always converges to a domain of inhibited traces $I_n$. We call such a $I_n$ a complete domain of inhibited traces for $S$ and $P$.

Now, to demonstrate the correctness of our proposal, we prove that an adaptor with a complete domain of inhibited traces $I_n$ always lead the interacting services to successful states of the contract $(\text{OK}_c)$ while avoiding locks.

**Proposition 2 (Correctness).** Given the initial states of two bounded services $S$ and $P$, an adaptor with an initial state $A_0$ with contract $c$. If the adaptor employs a monotonic learning function, and $I$ is a complete domain of inhibited traces, then for every $S', A', t'$ and $P'$ such that $S | \langle A_0, I, \lambda \rangle_c[P] \xrightarrow{\tau}^* S' | \langle A', I, t' \rangle_c[P']$

where $A' \neq A_0$, there exists a sequence of $\tau$ transitions $S' | \langle A', I, t' \rangle_c[P'] \xrightarrow{\tau}^* S'' | \langle A'', I, t'' \rangle_c[P'']$

such that $A'' \cap \text{OK}_c \neq \emptyset$.

This result is particularly interesting in those cases where the adaptation contract guarantees that the services have successfully finished, i.e., those in which $S''$ and $P''$ are also final states of their respective services. This happens in our running example because the contract automaton (Fig. 1(b)) is aware of the ending of the services due to correspondence rules $v_c, v_e$ and $v_f$.

It is worth noting that the sequence $\{I_i\}_{i \in \{0, \ldots, n\}}$ of inhibited traces derived from Proposition 1 could be different for each run-time session. In this way, different learning iterations may lead to different complete domains of inhibited traces. Thus, we need to prove that the learning process is well defined, in the sense that the learning process does not depend on the execution or, in other words, the complete domains of inhibited traces are “essentially” the same. The following proposition illustrates this result.
Proposition 3 (Well-definedness). Let $S$ and $P$ be the initial states of two bounded services. Let us consider an adaptation contract $c$ which corresponds to an adaptor with an initial state $A_0$ and a monotonic learning function. If $I$ and $I'$ are complete domains of inhibited traces resulting from a learning process starting in $S \parallel \langle A_0, I_0, \lambda \rangle_c [P]$, then

$$I \subseteq I' \quad \text{and} \quad I' \subseteq I$$

4 Learning Policies

We now show how different definitions of $\text{add}(t, I)$ can be employed to define different learning policies for rule (Learn).

Bounded learning. An upper bound to the number of traces that are inhibited by an adaptor at any given time may be set for different reasons. The most common is memory capacity, which may limit the size of learned information that can be kept in memory. To respect such a limit, adaptors may need to forget some previously inhibited traces when learning a new trace to be inhibited. A simple bounded learning policy is to forget (if needed) the oldest learned trace when learning a new one. For this, we are going to model inhibited traces $I$ as a sequence of traces (e.g., $I = t_0; \cdots; t_n$) where $I \vdash t$ iff $I = I':t';$

$$\text{add}_b(t, I) = \begin{cases} J : t & \text{if } \text{outOfBound}(I : t, \beta) \text{ and } I = u.J \\ I : t & \text{otherwise} \end{cases}$$

where $\text{outOfBound}(I : t, \beta)$ holds if the size of $I : t$ exceeds the maximum allowed size $\beta$. Other types of bounded learning policies can be implemented by defining different $\text{outOfBound}$ boundedness conditions (e.g., on the number of traces — rather than on their size) and/or by choosing differently which trace(s) to forget (e.g., one of the longest traces — rather than the oldest one). For instance:

$$\text{add}_b(t, I) = \begin{cases} \text{del}^b(D, I) & \text{if } \text{outOfBound}(I : t, \beta) \text{ and } I = a(D, J) \\ I : t & \text{if } I = a(u, J) \text{ and } D \vdash u \\ \emptyset & \text{if } I = \emptyset \end{cases}$$

where $\text{ap}(t, I) = I : t$; $\text{longest}(I) = \{u \mid I \vdash u \wedge \exists t . |t| > |u|\}$; and $\text{del}$ is recursively defined as follows:

$$\text{del}^b(D, I) = \begin{cases} \text{del}^b(D, J) & \text{if } I = a(u, J) \text{ and } D \vdash u \\ a(u, \text{del}^b(D, J)) & \text{if } I = a(u, J) \text{ and } D \vdash u \\ \emptyset & \text{if } I = \emptyset \end{cases}$$

Prefix-driven absorption. The way in which adaptors forget inhibited traces affects the overall performance of learning adaptors as much as the way in which they learn them. While bounded learning policies indirectly define a (boundedness determined) forget policy, trace prefixing can be exploited to

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4 Since boundedness conditions are often application- and device-dependent, bounded learning policies are parameterised w.r.t the maximum allowed size $\beta$. 

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intentionally define a forget policy to shrink the size of learned information. Intuitively speaking, the inhibition of a trace \( t \) which is a prefix of a previously inhibited trace \( t \vdash u \) subsumes (by rules (Ext) and (Int)) the inhibition of the latter, which hence does not need to be explicitly stored among the inhibited traces anymore. A learning policy based on prefix-driven absorption can be easily specified by defining \( \text{add}(t, I) \) as:

\[
\text{add}^2(t, I) = a(t, \text{del}^a(\text{prefixedBy}(t, I), I))
\]

where \( \text{prefixedBy}(t, I) = \{ u \mid I \vdash u \land \exists v. u = t \vdash v \} \) is the set of traces in \( I \) that are prefixed by \( t \). It is worth observing that different learning policies can be combined together. For instance, prefix-driven absorption and bounded learning policies can be naturally combined into a single policy as follows.

\[
\text{add}^2_{2+1}(t, I) = \text{add}^2_{a+1}(t, I)
\]

It is also worth observing that prefix-driven absorption can also be exploited to identify temporary failures not due to service protocol incompatibilities. To do that we must distinguish “simple” prefixes from “non-simple” prefixes. We say that \( t \) is a simple prefix of \( t \vdash u \) if \( u \) contains only one element. The normal learning process of an adaptor may inhibit a simple prefix \( t \) of a previously inhibited trace \( t : a \) whenever the adaptor realises that there is no alternative extension of \( t \). On the other hand, the inhibition of a trace \( t \) which is a non-simple prefix of a previously inhibited trace \( t \vdash u \) might be caused by some temporary failure that intervened (e.g., physical communication problems —such as shadow fading or increased physical distance). The detection of temporary failures can be exploited to define refined prefix-driven absorption policies that maintain the set \( I_T \) of traces learned from temporary failures separate from the set \( I_P \) of traces learned from (supposedly) permanent failures\(^5\), such as the following definition of \( \text{add} \). Let \( \text{onePrefixedBy}(t, I) = \{ t \vdash \alpha \mid I \vdash t \vdash \alpha \} \) be the set of traces \( t : \alpha \) in \( I \) which are prefixed by \( t \) and which are only one element longer than \( t \). Then:

\[
\text{add}^3(t, \langle I_P, I_T \rangle) = \begin{cases} 
(a(t, \text{del}^a(J, I_P)), \text{del}^a(J, I_T)) & \text{if onePrefixedBy}(t, I_P) \neq \emptyset \text{ and } J = \text{prefixedBy}(t, I_P \cup I_T) \\
(\text{del}^a(J, I_P), a(t, \text{del}^a(J, I_T))) & \text{if onePrefixedBy}(t, I_P) = \emptyset \text{ and } \text{prefixedBy}(t, I_P \cup I_T) = J \text{ and } J \neq \emptyset \\
(a(t, I_P), I_T) & \text{otherwise}
\end{cases}
\]

Combined policies whose bounded learning and/or time-to-forget components prioritarily forget traces corresponding to temporary failures can be easily defined. For instance, let \( \langle I_P, I_T \rangle = \text{add}^3_{a+1}(t, \langle I_P, I_T \rangle) \), then:

\[
\text{add}^3_{a+1}(t, \langle I_P, I_T \rangle) = \begin{cases} 
\langle I_P, J_T \rangle & \text{if outOfBound}(I_P :: I_T, \beta) \text{ and } I_P = u.J_T \\
\langle J_P, \emptyset \rangle & \text{if outOfBound}(I_P :: I_T, \beta) \text{ and } I_T = \emptyset \text{ and } I_P = u.J_P \\
\langle I_P, I_T \rangle & \text{otherwise}
\end{cases}
\]

\(^5\) Rules (Ext) and (Int) trivially extend to the case in which \( I \) is modelled as a pair \( (I_P, I_T) \), viz., by turning \( I \not\models t \) into \( I_P \cup I_T \not\models t \).
Reset on empty adaptors. The aforementioned learning policies aim at reducing the memory requirements (add1 and add2) and mitigate sporadic errors (add3). In particular, the main problem of the basic learning policy (add0) with sporadic errors (unforeseen failures in the synchronisations due to instabilities in the communication channels) is that it tends to converge to the empty adaptor. This happens because add0 does not forget the inhibited traces due to this sporadic errors and, as a result, the adaptor behaviour is constantly reduced every time one of these errors occurs. An straightforward solution to this issue, is to recognise when the process has converged to the empty adaptor and then reset the inhibited traces so that the adaptor can converge to better solutions. This is formalised with the following function.

\[ add^a_4(t, I) = \begin{cases} \emptyset & \text{if } t = \lambda \\ a(t, I) & \text{otherwise} \end{cases} \]

Intuitively, add4 behaves as a when the adaptor is not empty. If it becomes empty, and this is not considered valid by the given contract, then the only rule that can be triggered is the rule Learn inhibiting the empty trace \( \lambda \) as far as no synchronisation is possible with the empty adaptor. When this happens, function add4 clears the inhibited traces so that the adaptor can synchronise again. As usual, add4 can be combined with other learning policies, e.g.:

\[ add_{a+2+1}(t, I) = add^{a_2+1}_4(t, I) \]

It is easy to prove that add_i, i ∈ \{0, 2, 3\} (and their combinations) are monotonic by Definition 2 whereas add1, add4 are not (deliberately). We will see in Sect. 5 that, although non-monotonic learning policies do not necessarily converge, they have the advantage of overcoming sporadic errors while maintaining high success rates.

5 Evaluation and Tool Support: ITACA

Learning adaptors have been implemented and included in the Integrated Toolbox for Automatic Composition and Adaptation (ITACA\(^6\) [4]). We have evaluated our approach with two real-world data-diffusion protocols for sensor networks: TinyDiffusion [14] and SPIN [8] (see Sect. B). In this experiment, a TinyDiffusion node was adapted to participate in the communication between two SPIN nodes. The example can be solved with a minimum of 55 inhibited traces (corresponding to 7123 adaptor transitions which do not need to be stored in memory) and it allows a maximum of 5466 different successful traces. We use a transition error rate parameter \( (TER \in [0, 1]) \) which represents the probability of a synchronisation to forcibly fail due to sporadic errors. The experiment is run 10 times to plot the arithmetic mean and the sample standard deviation.

\(^6\) http://itaca.gisum.uma.es/
Different learning policies are compared in Fig. 4, where the number of simulated traces is shown in the horizontal axis. Line \textit{reg} corresponds to a regular adaptor using \textit{add}_2. Line \textit{dthr} represents an adaptor using \textit{add}_{4+2+1} with a dynamic threshold $\beta \in \mathbb{N}$, initially set to 0, which is incremented and decremented each time rules \textit{Ok} and \textit{Learn} are respectively used. The adaptive adaptor, \textit{athr}, also uses \textit{add}_{4+2+1} with a dynamic threshold $\beta' \in \mathbb{N}$ but, in this case, $\beta'$ is always set to be equal to the number of transitions in the adaptor. Finally, \textit{noi} represents an adaptor which does not learn, i.e., $I$ is always empty. The latter is used as a comparative baseline for the other approaches.

Figure 4(a) shows the success rate, i.e., the percentage of simulated traces which were successful in the current interval. It can be seen that \textit{noi} remains close to a success rate of 55%, which is reduced proportionally to the TER. Adapter \textit{dthr} performs slightly better, but not significantly due to its low threshold $\beta$. The other adaptors take advantage of the learning process and achieve success rates close to 100%. However, when sporadic errors start to occur (starting from simulation 4000), adaptor \textit{reg}, which is not able to forget inhibited traces, quickly converges to the empty adaptor and remains so for the rest of the simulation. Finally, \textit{athr} is also affected by high values of TER but it is able to recover when sporadic errors cease to occur, achieving success rates close to 100%.

A detail of the \textit{athr} adaptor is depicted in Fig. 4(b). It shows the amount of inhibited traces ($I$), sporadic errors ($E$) and the total number of failed traces ($F \geq E$). The number of inhibited traces initially approximates the desired value of 55. However, when sporadic errors appear (4000), new inhibited traces reduce the size of the adaptor (i.e., number of transitions), this reduces threshold $\beta'$ which finally reduces the number of inhibited traces. Intuitively, this means that the adaptor reduces its knowledge because it cannot trust it. This phenomenon reappears when TER is increased in subsequent iterations (6000, 8000, 10000 and 12000). The final range (14000, 20000] is more interesting. We can see that, although \textit{athr} succeeds in recovering from sporadic errors, achieving success rates close to 100%, it does so at the cost of obtaining a suboptimal, but correct, solution. In other words, depending on where the sporadic errors occurred, adaptor \textit{athr} might prune bigger parts of the behaviour than what is needed.

Interestingly, adaptor \textit{reg} enhanced with reset capabilities as \textit{dthr} (i.e., \textit{reg+reset} using \textit{add}_{4+2}) was able to match \textit{athr}.\footnote{The statistics characterising \textit{reg+reset} are indistinguishable from those of \textit{athr}.} This fact leads to the conclusion that it is not the dynamic threshold what matters but to be able to notice the convergence to empty adaptor, and thus reset the inhibited traces. Therefore, the most promising adaptor is \textit{reg+reset} (\textit{add}_{4+2}) thanks to its simplicity and effectiveness.

Regarding the computational complexity, every synchronisation with the adaptor requires a transition in the adaptor behaviour and the possible inclusion of a new inhibited trace. The pseudocode for this algorithm is shown in Appendix C. Assuming hash sets and hash maps with constant complexity for membership queries and insertions, the time complexity is $O(|S^c||\Sigma^c|^4)$ where $|S^c|$ is the number of states in the contract automaton, $|\Sigma^c|$ is the number of correspondence

\footnote{The statistics characterising \textit{reg+reset} are indistinguishable from those of \textit{athr}.}
rules in the contract and \( l \) is the maximum length of a trace. The spatial complexity of our approach with \( \text{add}_i, i \in \{0, 2, 3, 4\} \) is given by the combined size of: the inhibited traces, the adaptor state and the adaptation contract. The space required by inhibited traces can be reduced either by storing them as a tree or using any learning policy based on \( \text{add}_1 \) (where the size of the inhibited traces is bounded by \( \beta \)). Both approaches result in a spatial complexity of \( O(|\mathcal{S}| \cdot |\Sigma|^{l}) \).

Both complexities are greatly reduced if the adaptation contract is deterministic in the sense that it does not require the lazy-choice represented by rule L. In this case, at any given adaptor state \( \langle A, I, t \rangle \), it happens that \( A \) contains a single element \( \langle s, \Delta \rangle \). This simplification results in a time complexity of \( O(\max(|\Sigma|, l)) \) and a spatial complexity of \( O(|\mathcal{S}| \cdot |\Sigma|^{l} + |\Sigma|^l) \).

6 Related Work

The behaviour of an adaptor can be synthesised at design time following other approaches covered in related work [1,4,11,13]. However, adaptor synthesis is exponential with regard to the number and size of the services involved and it requires to know in advance the behaviour of these services. This is not feasible in the current setting of nodes with restricted capabilities where the actual behaviour of the services is unknown. For instance, the correspondence rules presented in this work are similar to the adaptation operators presented in [6] and to the mismatch patterns introduced in [11], but their approaches are focussed on design-time. So there are few related work which aim at addressing both runtime and lightweight behavioural adaptation at the same time.

One of them is [5], where an ontology is required to generate a mapping between the operation of the services. Some properties (expressed in a temporal logic) are dynamically verified by performing forward-search analyses on the behaviour of services. While similar properties can be encoded with our adaptation
contract automata, differently from us, \[5\] requires the behaviour of services to be known and it has to bear with the cost of the forward-search analysis.

Wang et al. \[16\] propose the dynamic application of adaptation rules. These rules are triggered by the input actions received by the adaptor and then an output action is generated. Our approach is similar to theirs in the sense that we also apply the adaptation contract dynamically without generating the whole adaptor. However, their rules must specify how to solve both signature and behavioural incompatibilities, hence requiring to know the behaviour of the services beforehand. Our contracts, instead, only specify the how to solve signature incompatibilities and an optional description of the adaptation goal. Then, our adaptors dynamically learn how to solve behavioural incompatibilities.

Another related work is \[9\], where the problem of controlling services with unknown behaviour is discussed. Intuitively, this work shares with our approach the idea of progressively refining an over-approximated controller when failures occur. The authors of \[9\] perform such refinement by exploiting (bounded) model checking, whose overhead is not bearable in applications running on limited capacity devices.

7 Conclusion

We have presented a new lightweight approach to behavioural runtime adaptation. Our approach requires an adaptation contract based on the signatures of the services (the collection of operations they require and offer), but no previous knowledge on the behaviour of the services is needed, which will be dynamically learned. We have shown how adaptors can incrementally learn from interaction failures at run time so as to eventually converge to the same behaviour that could be a priori synthesised by means of (computationally expensive) analyses on the behaviour of the services at design time.

The learning adaptors presented in this work can be applied to perform zero-knowledge adaptation, i.e., adaptors without adaptation contract. In this case, there is an implicit contract which assumes that every source and destination service share the same alphabet of actions, therefore presenting a trivial set of one-to-one correspondence rules. Having such a zero-knowledge contract, which is dynamically inferred, the adaptor does not perform any adaptation at signature level (it simply forwards messages), but it does learn from possible behavioural incompatibilities between the services (such as message expected in different order), therefore it avoids the deadlocks that would be present without adaptation.

As future work, we will investigate process mining techniques to improve the efficiency of the approach.

References

Analyzing Reconfigurable Component-Based Systems Using Attribute Grammars*

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Abstract. Reconfigurable systems have pervaded many fields of computing and are becoming more and more important. To make them amenable to systematic design and analysis methods, formal modeling languages are required. One such language is AADL (Architecture Analysis and Design Language), which is gaining widespread acceptance in automobile, avionics and aerospace industries for comprehensively modeling component-based, safety-critical distributed systems by capturing functional, probabilistic and hybrid aspects. In particular, AADL supports the specification of dynamic reconfiguration of systems in the form of mode transitions which (de-) activate components and their communication connections. However it turns out that one has to use this mechanism carefully as it can give rise to cyclic dependencies between communication ports, which are forbidden.

In this paper we show that the problem of cyclic data port dependencies in AADL specifications is closely related to the circularity problem in attribute grammars, a well-known formalism for describing syntax-directed translations such as the semantic analysis or code generation in compilers for programming languages. Exploiting this relation, we are able to reuse existing circularity tests for analyzing AADL specifications.

Keywords: Reconfigurable systems, AADL, Port dependencies, Attribute grammars, Circularity

1 Introduction

Reconfigurable systems have pervaded many fields of computing and are becoming more and more important. This is particularly true for aerospace applications where maximizing the potential of high-performance payload design against multiple missions requires both performance and flexibility. To make reconfigurable systems amenable to systematic design and analysis methods, formal modeling languages are required. One such language is AADL (Architecture Analysis and Design Language), which is gaining widespread acceptance in automobile, avionics and aerospace industries for comprehensively modeling safety-critical distributed systems by capturing functional, probabilistic and hybrid aspects.

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The central concept in AADL is that of a component. Systems are hierarchically organized, i.e., components can contain other components, entailing the notion of subcomponents. The operational state of a component is determined by its current mode, and its behavior is defined by transitions between modes which are triggered by incoming events, or which generate outgoing events. Components can exchange values via data ports whose interconnection structure is specified in the component implementation by introducing connections (between components) and flows (within a component).

With regard to dynamic reconfiguration, the key feature of AADL is that the activation of subcomponents, connections and flows can be made dependent on the current mode of the respective component. In other words, a mode transition of that component can activate or deactivate some of its subcomponents and change their interconnection structure. However it turns out that one has to use this dynamic reconfiguration mechanism carefully as it can give rise to cyclic dependencies between data ports. Such cyclic dependencies are disallowed for semantic reasons as they give rise to recursive equations on data port values, which are not (uniquely) solvable in general. What is required therefore is a checking procedure that tests whether, in any mode configuration of the overall system, there exists a cycle in the dependency graph over the data ports that is induced by the connections and flows that are active in that configuration.

In this paper we show that this problem is closely related to the circularity problem in attribute grammars, a well-known formalism for describing syntax-directed translations such as the semantic analysis or code generation in compilers for programming languages. Exploiting this relation, we are able to reuse the circularity tests that have been developed in the past for analyzing AADL specifications. We also sketch the implementation of one such test in a toolset for the analysis and verification of AADL specifications.

The remainder of this paper is organized as follows. Section 2 gives a brief introduction to AADL, with a particular emphasis on the specification of reconfigurable systems. Section 3 explains the essential concepts of attribute grammars and the circularity problem. Section 4 presents the main result of this paper, the relation between (the cyclicity problem of) AADL specifications and (the circularity problem of) attribute grammars. Section 5 briefly describes a toolset that implements the corresponding test. Finally, Section 6 presents some conclusions.

2 Specifying Reconfigurable Systems in AADL

2.1 Overview of AADL

Developed by a Society of Automotive Engineers (SAE) sponsored committee of experts, the Architecture Analysis and Design Language (AADL) was approved and published as SAE Standard AS-5506 in 2004 [11]. Models in this language can capture real-time, performance critical and distributed aspects of a system at an architectural level. It supports a component-based, model-driven development approach throughout the system life cycle. More concretely, AADL has the following design features:
Modeling both the system’s nominal and faulty behavior. To this aim, AADL (and its Error Model Annex [12]) provides primitives to describe software and hardware faults, error propagation (that is, turning fault occurrences into failure events), sporadic (transient) and permanent faults, and degraded modes of operation (by mapping failures from architectural to service level).

Specifying timed and hybrid behavior. In particular, in order to analyze continuous physical systems such as mechanics and hydraulics, AADL supports continuous real-valued variables with (linear) time-dependent dynamics.

Modeling probabilistic aspects, such as random faults, repairs, and stochastic timing.

A complete specification consists of three parts, namely a description of the nominal behavior, a description of the error behavior, and a fault injection specification that describes how the error behavior influences the nominal behavior. In this paper we concentrate on the nominal specification.

The nominal model describes the system under normal operation. It is a system decomposition of interacting components in which system details can be abstracted by defining a hierarchy among components. The interaction interfaces are specified using port connections of which there are three types. Data port connections expose port values to other components, flow port connections (flows for short) are evaluable functions based on incoming data ports, and event port connections are used to define (multi-way) hand-shaking communication between components. These port interfaces are complemented by a mode transition system, which describes changes over the ports, and thus essentially captures a component’s behavior. The transition system can be annotated with linear differential equations and timing constraints to model the evolution of physical aspects, like temperature, pressure and scheduling of tasks. Modes can also be used to represent degradation of the system. Transitions between these modes can lead to dynamic reconfigurations by (de-)activating components and port connections, as shown in the following subsection.

2.2 Language Features for Specifying Dynamic Reconfiguration

As mentioned earlier, dynamic reconfiguration of systems is supported in AADL by (de-)activating components and connections. More concretely, the presence of subcomponents and port connections can be made dependent on the current mode of the respective component. Whenever a mode transition adds or removes a subcomponent, connection, or flow, we say that the former activates or deactivates the latter, respectively.

We introduce these features by means of a simple example, a redundant data acquisition system with main component Acquisition. As shown in Figure 1, it is comprised of a redundant sensor component, a redundant filter component, and a monitor component. Excerpts of the corresponding specification are given in Figures 2 and 3. The idea is that measurement values are captured by the currently active sensor. Then they are forwarded along the out-to-in data port connection between Sensors and Filters to the currently active filter, where
they are processed (in this example, simply doubled) and passed on to the outgoing data port value of the main component.

The output of both the sensor and the filter component is supervised by a monitor component, which triggers the switch to the spare component (using the respective switch event) if the output does not lie in the expected range. As can be seen, the activation status of components and connections is determined by the in modes clauses. If they are absent, the object is active in every mode of the respective supercomponent (such as subcomponent Sensors of supercomponent Acquisition).

Note that there are three different types of data port connections:

- from an incoming port of a supercomponent to an incoming port of one of its subcomponents, such as

  data port input -> filter1.input

  in Filters;

- from an outgoing port of a component to an incoming port of one of its neighbor components, such as

  data port sensors.output -> filters.input

  in Acquisition; and

- from an outgoing port of a subcomponent to an outgoing port of its supercomponent, such as

  data port filters.output -> value

  in Acquisition.

In particular, data port connections from incoming to outgoing ports are excluded. Such dependencies have to be defined using flows; please refer to the definition of Filter for an example.

### 2.3 Cyclic Dependencies between Data Ports

Data port connections and flows induce an instantaneous exchange of information. It must be ensured that in every reachable configuration of the system (that is, for every combination of component modes that occurs in some execution), every data port can be assigned a unique value. It must therefore always be guaranteed that the dependency relation on the collection of all data ports of the system which is imposed by flows and data port connections is acyclic.

In order to make this notion precise, we use the following formal description of an AADL specification.

- The set of components is denoted by $Cmp$, and the main component is denoted by $main$. 


For each component $c \in Cmp$, the (finite) set of its modes is given by $Mod(c)$, and the set of its (direct) subcomponents which are active in the respective mode $m \in Mod(c)$ (according to the in modes clauses) is denoted by $Act(c, m) \subseteq Cmp$. If the component does not have an associated mode, we assume an implicitly defined default mode defmod.

The incoming and outgoing data ports of $c \in Cmp$ are collected in the (disjoint) sets $InPrt(c)$ and $OutPrt(c)$, respectively, and we let $Prt(c) := InPrt(c) \cup OutPrt(c)$.

As described earlier, data port connections fall into three different categories, depending on the direction of the ports (in or out) and the relation of the participating components (super or sub). To obtain a concise notation, we use the index 0 to refer to the respective supercomponent, and positive indices for its direct subcomponents. Formally, for $c_0 \in Cmp$, $m \in Mod(c_0)$, and $Act(c_0, m) = \{c_1, \ldots, c_n\}$,

$$Con(c, m) \subseteq \{(p, i, q, j) \mid i = 0, j \in [n], p \in InPrt(c_0), q \in InPrt(c_j), \text{ or }$$

$$i, j \in [n], p \in OutPrt(c_i), q \in InPrt(c_j), \text{ or }$$

$$i \in [n], j = 0, p \in OutPrt(c_i), q \in OutPrt(c_0)\}$$

where $[n]$ abbreviates $\{1, \ldots, n\}$.
system Acquisition
features
value: out data port real;
end Acquisition;
system implementation Acquisition.Impl
subcomponents
sensors: system Sensors;
filters: system Filters;
monitor: system Monitor;
connections
data port sensors.output -> filters.input;
data port sensors.output -> monitor.valueS;
data port filters.output -> value;
data port filters.output -> monitor.valueF;
event port monitor.switchS -> sensors.switch;
event port monitor.switchF -> filters.switch;
end Acquisition.Impl;

system Sensors
features
output: out data port real;
switch: in event port;
end Sensors;
system implementation Sensors.Impl
subcomponents
sensor1: device Sensor in modes (Primary);
sensor2: device Sensor in modes (Backup);
connections
data port sensor1.output -> output in modes (Primary);
data port sensor2.output -> output in modes (Backup);
modes
Primary: initial mode;
Backup: mode;
transitions
Primary -[switch] -> Backup;
end Sensors.Impl;
device Sensor
features
output: out data port real;
end Sensor;
device implementation Sensor.Impl
...
end Sensor.Impl;

Fig. 2. Specification of main and sensor components (excerpt).
system Filters
features
  input: in data port real;
  output: out data port real;
  switch: in event port;
end Filters;

system implementation Filters.Impl
subcomponents
  filter1: device Filter in modes (Primary);
  filter2: device Filter in modes (Backup);
connections
  data port input -> filter1.input in modes (Primary);
  data port input -> filter2.input in modes (Backup);
  data port filter1.output -> output in modes (Primary);
  data port filter2.output -> output in modes (Backup);

modes
  Primary: initial mode;
  Backup: mode;
transitions
  Primary –[switch]– Backup;
end Filters.Impl;

device Filter
features
  input: in data port real;
  output: out data port real;
end Filter;

device implementation Filter.Impl
flows
  output := 2.0 * input;
end Filter.Impl;

system Monitor
features
  valueS: in data port real;
  valueF: in data port real;
  switchS: out event port;
  switchF: out event port;
end Monitor;

system implementation Monitor.Impl
modes
  check: initial mode;
transitions
  check –[switchS when ...valueS...]-> check;
  check –[switchF when ...valueF...]-> check;
end Monitor.Impl;

Fig. 3. Specification of filter and monitor components (excerpt).
Finally, for \( c \in \text{Cmp} \) and \( m \in \text{Mod}(c) \) the flows of \( c \) that are active in \( m \) are denoted by

\[
\text{Flw}(c, m) \subseteq \{(e, q) \mid e \text{ expression over } \text{InPrt}(c), q \in \text{OutPrt}(c)\}.
\]

For example, the specification of the data acquisition system in Figures 2 and 3 is formalized as follows.

- \( \text{Cmp} = \{ \text{Acquisition}, \text{Sensors}, \text{Sensor}, \text{Filters}, \text{Filter}, \text{Monitor} \} \)
- \( \text{main} = \text{Acquisition} \)
- \( \text{Mod}(\text{Acquisition}) = \{ \text{defmod} \} \)
  \( \text{Act}(\text{Acquisition}, \text{defmod}) = \{ \text{Sensors}, \text{Filters}, \text{Monitor} \} \)
  \( \text{Mod}(\text{Sensors}) = \{ \text{Primary}, \text{Backup} \} \)
  \( \text{Act}(\text{Sensors}, \text{Primary}) = \{ \text{Sensor} \} \)
  \( \text{Act}(\text{Sensors}, \text{Backup}) = \{ \text{Sensor} \} \)
- \( \text{InPrt}(\text{Acquisition}) = \emptyset \)
  \( \text{OutPrt}(\text{Acquisition}) = \{ \text{value} \} \)
  \( \text{InPrt}(\text{Filters}) = \{ \text{input} \} \)
  \( \text{OutPrt}(\text{Filters}) = \{ \text{output} \} \)
- \( \text{Con}(\text{Acquisition}, \text{defmod}) = \{(\text{output}.1, \text{input}.2), \) 
  \( \quad (\text{output}.1, \text{valueS}.3), \) 
  \( \quad (\text{output}.2, \text{value}.0), \) 
  \( \quad (\text{output}.2, \text{valueF}.3)\}\}
  \( \text{Con}(\text{Sensors}, \text{Primary}) = \{(\text{output}.1, \text{output}.0)\} \)
  \( \text{Con}(\text{Sensors}, \text{Backup}) = \{(\text{output}.2, \text{output}.0)\} \)
- \( \text{Flw}(\text{Acquisition}, \text{defmod}) = \emptyset \)
  \( \text{Flw}(\text{Filter}, \text{defmod}) = \{(2.0 \ast \text{input}, \text{output})\} \)

Note that \( \text{Con} \) and \( \text{Flw} \) represent dependencies between data ports that are induced by connections and flows, respectively, on the level of a single component. On this level, cyclic dependencies are excluded since outgoing data ports of the respective component cannot be connected to incoming ports; this can only happen in a larger context. Thus we have to lift component-level dependencies to system-level dependencies by considering all possible system configurations, i.e., assignments of modes to components. For each configuration, the corresponding dependency graph can be obtained by concatenating the dependency graphs of all active components in their respective mode. For example, Figure 1 visualizes the data port dependencies of the data acquisition system in a configuration that uses the primary sensor and the primary filter.

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Under these premises, an AADL specification is called cyclic if there exists a mode configuration such that the corresponding data port dependency graph has a (directed) cycle.

It is important to notice that this property is undecidable if we restrict it to reachable configurations, i.e., to those that can occur in any concrete execution of the system. As the AADL fragment for describing mode transitions is quite expressive (involving computations and tests on integers, for example), the reachability problem for configurations is undecidable and therefore also the corresponding cyclicity problem. We thus have to approximate it by considering all combinations of modes, no matter whether they are reachable or not. More formal details on the formalization of AADL specifications can be found in [1, 2].

Also note that data port connections alone are not sufficient to yield a dependency cycle, as connections are only involving ports of the same direction, or are running from output to input ports. In order to close a cycle, it must be possible to introduce a dependency from an input to an input port, which requires a flow.

As can be seen in Figure 1, the given system configuration is cycle-free. In fact, this applies to every configuration of the system. This would be different if, for example, we introduced a flow in the Monitor component that processes the value information coming from the Filters component and feeds it back to the input port of that component along a connection using a new outgoing data port.

3 Attribute Grammars

Attribute grammars were devised by Knuth [8, 9] to describe semantic aspects of context-free languages. They give a meaning to every derivation tree of the underlying context-free grammar by assigning values to various attributes associated with the nodes of the tree. Nowadays they are mainly used for implementing syntax-directed translations such as the semantic analysis or code generation in compilers for high-level programming languages [5, Chapter 3].

3.1 Definition of Attribute Grammars

We start with the definition of a context-free grammar, which is of the form $G = \langle N, \Sigma, P, S \rangle$ where

- $N$ is a finite set of nonterminal symbols;
- $\Sigma$ is a (finite) alphabet of terminal symbols (disjoint from $N$);
- $P$ is a finite set of production rules of the form $A \rightarrow \alpha$ where $A \in N$ and $\alpha \in X^*$ for $X := N \cup \Sigma$; and
- $S \in N$ is a start symbol.

A context-free grammar is extended to an attribute grammar by attaching attributes to the nonterminal symbols. These attributes come in two variants:
- **inherited** attributes, which are used for passing context information top-down from the root towards the leaves; and
- **synthesized** attributes, which are used for collecting information in a subtree and for passing it bottom-up towards the root.

This information flow is realized in the following way. In the context of each production, the values of the synthesized attributes of the left-hand side nonterminal (for the bottom-up flow) and the values of the inherited attributes of the right-hand side nonterminals (for the top-down flow) have to be defined by a so-called **semantic rule**. The latter is an equation whose right-hand side may refer to the complementary set of attributes, that is, to the inherited attributes of the left-hand side nonterminal and the synthesized attributes of the right-hand side nonterminals.

Formally this can be described by a tuple of the form $\mathfrak{A} = \langle N, Atr, atr, D, E \rangle$ with the following components.

- $Atr = Inh \cup Syn$ is the set of (respectively inherited and synthesized) attributes.
- The function $atr : N \rightarrow 2^{Atr}$ denotes the attribute assignment, and we let $inh(A) := atr(A) \cap Inh$ and $syn(A) := atr(A) \cap Syn$ for every $A \in N$.
- $D := \{D_\alpha \mid \alpha \in Atr\}$ is the family of domains, which restrict the possible values of each attribute in $Atr$.
- Every production $\pi = Y_0 \rightarrow Y_1 \ldots Y_n \in P$ determines the set
  \[ Var_\pi := \{ \alpha.i \mid i \in \{0, \ldots, n\}, Y_i \in N, \alpha \in atr(Y_i) \} \]
  of attribute variables of $\pi$. It is decomposed into the subsets of **inner and outer variables**
  \[ In_\pi := \{ \alpha.i \mid i = 0, \alpha \in syn(Y_i) \text{ or } i \in [n], \alpha \in inh(Y_i) \} \]
  \[ Out_\pi := Var_\pi \setminus In_\pi \]
  which are respectively used on the left-hand and right-hand sides of semantic rules.
- A **semantic rule** of $\pi$ is an equation of the form
  \[ \alpha.i = f(\alpha_{i_1}.i_{j_1}, \ldots, \alpha_{i_k}.i_k) \]
  where $k \in \mathbb{N}$, $\alpha.i \in In_\pi$, $\alpha_{i_j}.i_j \in Out_\pi$ ($j \in [k]$), and $f : D_{\alpha_{i_1}} \times \ldots \times D_{\alpha_{i_k}} \rightarrow D_\alpha$.
- For each $\pi \in P$, $E_\pi$ is a set with exactly one semantic rule for every inner variable of $\pi$, and we let $E := \{ E_\pi \mid \pi \in P \}$.

For example, the attribute grammar given in Figure 4, taken from [8], specifies the evaluation of a given binary number (with fraction). It employs the synthesized attribute $v$ (of nonterminals $N$, $L$, and $B$) to compute the respective value in $D_v := \mathbb{Q}$. Moreover $l \in Syn$ represents the length of a bit list $L$ ($D_l := \mathbb{N}$), and $p \in Inh$ stands for the position of a single bit $B$ ($D_p := \mathbb{Z}$).

In general, an attribute grammar is used in the following way. To compute the semantic value of a given input word, say $w$,
1. the corresponding derivation tree with respect to the underlying context-free grammar is constructed; and
2. the attribute occurrences are evaluated according to the semantic rules.

To implement the second step, several methods have been proposed in the literature (see [4] for an overview). For example, the result of evaluating all attribute occurrences for the input word 10.1 using the attribute grammar from Figure 4 is shown in Figure 5. The latter also visualizes the corresponding attribute dependencies by arrows.

Just as cyclic data port dependencies in AADL specifications are critical, attribute evaluators usually cannot cope with cyclic dependencies between attribute occurrences, as the associated attribute equation system generally does not have a unique solution in this case. In the context of attribute grammars, this situation is called circularity.

3.2 (Strong) Circularity

Just as concatenating data port connections and flows for all components of the given AADL specification can give rise to cyclic dependencies between data ports, the same is true for attribute grammars.

An attribute grammar is called *circular* if there exists a syntax tree such that the corresponding attribute equation system is recursive (i.e., some attribute variable depends on itself).

Although a context-free grammar generally yields infinitely many derivation trees, the circularity problem for a given attribute grammar is nevertheless decidable. The corresponding algorithms (such as [10]) exploit the fact that, for every nonterminal symbol, there are only finitely many possible combinations of dependencies between the inherited and synthesized attributes of that symbol. However, one is faced with the problem of combinatorial explosion. The

---

Fig. 4. Attribute grammar for evaluating binary numbers.

| Numbers: | $N \rightarrow L$ | $v.0 = v.1$ |
|          | $p.1 = 0$         |
|          | $N \rightarrow L.L$ | $v.0 = v.1 + v.3$ |
|          | $p.1 = 0$         |
|          | $p.3 = -l.3$      |

| Lists:   | $L \rightarrow B$ | $v.0 = v.1$ |
|          | $l.0 = 1$          |
|          | $p.1 = p.0$        |
|          | $L \rightarrow LB$ | $v.0 = v.1 + v.2$ |
|          | $l.0 = l.1 + 1$    |
|          | $p.1 = p.0 + 1$    |
|          | $p.2 = p.0$        |

| Bits:    | $B \rightarrow 0$ | $v.0 = 0$ |
|          | $B \rightarrow 1$ | $v.0 = 2^{p.0}$ |
underlying reason is that for deriving the possible attribute dependencies on the left-hand side of a production, all combinations of dependencies for the right-hand side nonterminals have to be considered. In fact, in [6] it is shown that the circularity problem for attribute grammars is of intrinsically exponential complexity.

One approach to overcome this problem is a simplification of non-circularity which is called strong non-circularity [7,3]. The idea is to merge all possible dependencies between the inherited and synthesized attributes of a nonterminal symbol without distinguishing different (types of) derivation trees with the respective nonterminal at the root. As different derivation trees can entail different dependencies between the root attributes, this may yield spurious dependency cycles. (Please refer to Section 5 for an example.) The question whether this over-approximation does not yield any (true or spurious) cycle is decidable in polynomial time. Clearly, every strongly non-circular attribute grammar is also non-circular.

4 Analyzing AADL Specifications Using Attribute Grammars

After having introduced both AADL and attribute grammars, we can continue with describing the relation of the two formalisms in greater detail. Concretely, with a given AADL specification we will associate an attribute grammar such that the latter is circular if and only if the former is cyclic. This connection will then allow us to reuse circularity tests on attribute grammars for analyzing the AADL specification with respect to cyclic port dependencies.
AADL Attribute grammars

<table>
<thead>
<tr>
<th>System configuration</th>
<th>Derivation tree</th>
</tr>
</thead>
<tbody>
<tr>
<td>Active component</td>
<td>Nonterminal symbol</td>
</tr>
<tr>
<td>Inactive component</td>
<td>Terminal symbol</td>
</tr>
<tr>
<td>Mode</td>
<td>Production</td>
</tr>
<tr>
<td>Incoming data port</td>
<td>Inherited attribute</td>
</tr>
<tr>
<td>Outgoing data port</td>
<td>Synthesized attribute</td>
</tr>
<tr>
<td>Flow/data port</td>
<td>Semantic rule</td>
</tr>
</tbody>
</table>

Table 1. Relation between AADL and attribute grammars

The relation between AADL and attribute grammars is summarized in Table 1. The key idea is the analogy between system configurations on the one side and derivation trees on the other side. Remember that a configuration is defined by the current mode of each of the components in the system. It determines the active components: the main component is always active, and the current mode of the main component yields its active subcomponents according to the **in modes** clause given with its declaration. Recursively applying this definition and interpreting active and inactive components as nonterminal and terminal symbols, respectively, we obtain a derivation tree of a context-free grammar whose productions correspond to the modes of the AADL specification.

This idea is formalized by the following definition of a context-free grammar $G = (N, \Sigma, P, S)$ for a given AADL specification with component set $\text{Cmp}$, modes association $\text{Mod}$, and component activity mapping $\text{Act}$.

- $N := \text{Cmp}$;
- $\Sigma := \text{Cmp}^\dagger$;
- $P := \{\pi_{c,m} \mid c \in \text{Cmp}, m \in \text{Mod}(c)\}$ where $\pi_{c,m} := c \rightarrow c'_1 \ldots c'_n$ for $\bigcup_{m \in \text{Mod}(c)} \text{Act}(c, m) = \{c_1, \ldots, c_n\}$ and $c'_i := c_i$ if $c_i \in \text{Act}(c, m)$ and $c'_i := c_i$ otherwise; and
- $S := \text{main}$.

Here $\text{Cmp}^\dagger$ introduces a second name for each component to indicate its inactivity. In examples, we will use names with initial uppercase/lowercase letters to denote active/inactive components, respectively.

A further remark: it is possible to introduce data components in AADL specifications, which act as local variables in component implementations. (This feature was not used in the data acquisition example.) In the attribute grammar representation, they actually correspond to another kind of terminal symbols. As they cannot be equipped with data ports, however, they can be ignored for the cyclicity check. We have therefore not considered them in the translation.

Now that the context-free grammar part is fixed, we can continue with the attribute system. The principle is quite simple: (incoming and outgoing) data ports correspond to (inherited and synthesized, respectively) attributes, and the connections and flows are represented as semantic rules. Formally, with a given AADL specification with incoming/outgoing data ports $\text{InPrt}/\text{OutPrt}$, data port
connections $\text{Con}$, and flows $\text{Flw}$, we can associate an attribute grammar $A = \langle N, Atr, atr, D, E \rangle$ as follows.

- $\text{Inh} := \bigcup_{c \in \text{Cmp}} \text{InPrt}(c)$;
- $\text{Syn} := \bigcup_{c \in \text{Cmp}} \text{OutPrt}(c)$;
- for every $c \in \text{Cmp}$, $\text{inh}(c) := \text{InPrt}(c)$;
- for every $c \in \text{Cmp}$, $\text{syn}(c) := \text{OutPrt}(c)$; and
- for every $c \in \text{Cmp}$ and $m \in \text{Mod}(c)$,

$$E_\pi := \{ q.j = p.i \mid (p.i, q.j) \in \text{Con}(c, m) \}$$
$$\cup \{ q.0 = e[p \mapsto p.0; p \in \text{InPrt}(c)] \mid (e, q) \in \text{Flw}(c, m) \}.$$ 

Note that data port connections, which are equipped with indices to distinguish between super- and subcomponents, can directly be translated to semantic rules. In contrast, flows, which just employ data port names without indices (as they only refer to a single component), have to be adapted by attaching the index 0 to yield the corresponding rule. Here the expression $e[p \mapsto p.0]$ is obtained from $e$ by replacing every occurrence of $p$ by $p.0$.

The result of this association is the following.

The AADL specification is cyclic if and only if the associated attribute grammar is circular.

For example, the AADL specification of the data acquisition system as defined in Figures 2 and 3 yields the attribute grammar in Figure 6. Remember that names starting with uppercase letters refer to active components (nonterminals) while lowercase names denote inactive components (terminals).

\begin{verbatim}
Acquisition -> Sensors Filters Monitor      input.2 = output.1
                          valueS.3 = output.1
                          value.0 = output.2
                          valueF.3 = output.2
Sensors  -> Sensor sensor                   output.0 = output.1
Sensors  -> sensor Sensor                   output.0 = output.2
Sensor    -> e
Filters   -> Filter filter                  input.1 = input.0
                          output.0 = output.1
Filters   -> filter Filter                  input.2 = input.0
                          output.0 = output.2
Filter    -> e
Monitor   -> e
\end{verbatim}

Fig. 6. Attribute grammar representation of data acquisition system.
5 Implementation

The circularity test for attribute grammars as sketched in Section 3 has been used as the basis for an algorithm that checks whether a given AADL specification is cyclic. More concretely, in order to avoid the exponential complexity of the general problem [6], we decided to consider strong non-circularity [7], which can be tested in polynomial time.

The (somewhat artificial) specification in Figure 7 exemplifies the difference. As component $Sub$ occurs twice and as it has two modes, the overall system has four configurations. As can be seen from Figure 8, it is non-cyclic as none of those exhibits a cyclic dependency between data ports, and so the corresponding context-free grammar is non-circular. The algorithm for checking strong non-circularity, however, merges all dependencies that are induced by the flows defined for $Sub$. It thus yields the dependency graph that is shown in Figure 9, and that clearly exhibits a cycle. The attribute grammar is therefore not strongly non-circular.

The algorithm for testing strong non-circularity has been implemented in a toolset for analyzing and verifying AADL specifications. The corresponding project is entitled COMPASS (COrectness, Modelling and Performance of Aerospace SyStems), and is funded by the European Space Agency (ESA). More information and the tool download page can be found at http://compass.informatik.rwth-aachen.de/.

Our experiences show that considering strong non-circularity is sufficient for this purpose. Only AADL pathological specifications are non-cyclic but fail to pass our test. In our industrial case studies involving (parts of the) control systems of satellites, not a single system description of this kind was identified.

6 Conclusions

In this paper we have established a strong connection between specifications of reconfigurable systems in AADL on the one side and attribute grammars on the other side. Exploiting the idea that the (de-)activation of components and their communication structures by mode transitions can be interpreted as the application of production rules of a context-free grammar, we were able to show that the cyclicity problem for AADL specifications can be reduced to the circularity problem of attribute grammars. This approach has been successfully implemented in a toolset for analyzing AADL specifications.

References

system Super
end Super;

system implementation Super.Impl
  subcomponents
    sub1: system Sub;
    sub2: system Sub;
  connections
    data port sub1.out1 -> sub2.in2;
    data port sub1.out2 -> sub2.in1;
    data port sub2.out1 -> sub1.in1;
    data port sub2.out2 -> sub1.in2;
end Super.Impl;

system Sub
  features
    in1: in data port int;
    in2: in data port int;
    out1: out data port int;
    out2: out data port int;
end Sub;

system implementation Sub.Impl
  flows
    out1 := in2 in modes (m0);
    out1 := 1 in modes (m1);
    out2 := 2 in modes (m0);
    out2 := in1 in modes (m1);
  modes
    m0: initial mode;
    m1: mode;
  transitions
    ...
end Sub.Impl;

Fig. 7. A non-cyclic AADL specification.

Fig. 8. Configurations without cyclic dependencies.
Fig. 9. Cyclic dependency after merging.


Verifying Safety of Fault-Tolerant Distributed Components

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Abstract. We show how to ensure correctness and fault-tolerance of distributed components by behavioural specification. We specify a system combining a simple distributed component application and a fault-tolerance mechanism. We choose to encode the most general and the most demanding kind of faults, byzantine failures, but only for some of the components of our system. With Byzantine failures a faulty process can have any behaviour, thus replication is the only convenient classical solution; this greatly increases the size of the system, and makes model-checking a challenge. Despite the simplicity of our application, full study of the overall behaviour of the combined system requires us putting together the specification for many features required by either the distributed application or the fault-tolerant protocol: our system encodes hierarchical component structure, asynchronous communication with futures, replication, group communication, an agreement protocol, and faulty components. The system we obtain is huge and we have proved its correctness by using at the same time data abstraction, compositional minimization, and distributed model-checking.

1 Introduction

Safety in distributed systems is a wide research area which needs to be tackled at several levels: from the safety of the execution platform, to the correctness of the communication protocols and to correctness of the distributed applications. This article aims at evaluating the adequacy of formal method techniques for the verification of real-size distributed applications. The objective tackled by this article is really challenging because the application we consider features several non-functional concerns which contribute to the explosion of the number of states that can be reached by the application. Indeed we choose to provide a model and prove properties for a distributed application featuring fault-tolerance similar to Byzantine fault tolerance (BFT).

Our work is placed in the context of component oriented programming. Indeed from a programming model point of view, components provide well-defined

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modularity, and easiness to compose large applications from the composition of basic blocks. Also components require the precise definition of interfaces through which the basic blocks cooperate, which is crucial for a precise design of an application, but also strongly helps the formal specification of the application. Our components also allow a hierarchical and modular design, better specifying the structure of the application. We choose GCM[2] as our component model because it is naturally adapted to distribution, hierarchy, and one-to-many communication, but also it provides reconfiguration capabilities which we want to consider in future works. GCM is an extension of the Fractal component models with support for deployment, scalability, autonomic behaviour, and asynchronous communication; it also shares a lot of similarities with SCA [3]. In the VerCors [8] platform, we provide tools for verifying the behaviour of such distributed component applications.

This paper shows how to specify the behaviour and to verify properties of distributed component applications with request queues, future proxies and group proxies, and one-to-many interfaces. To illustrate our approach, we choose a simple distributed application featuring fault-tolerance by replication. Though the fault-tolerance properties we address are not outstanding, we think this application is a good opportunity to investigate on the use of model-checking to ensure safety of fault-tolerant applications. This article has the following objectives:

– Promote the use of formal methods to ensure safety of distributed systems.
– Provide a model for one-to-many communication.
– Study the modelling of faulty processes, and investigate the use of model-checking for verifying fault-tolerance from an application point of view. Indeed, most of the existing studies on this domain focus on the proof of correctness of the protocols only, not on the whole distributed application [14].
– Investigate the adequacy of distributed model-checking for verifying a distributed and asynchronous application that generates a huge state-space.

We do not model reconfiguration and adaptation, but we design our specification in such a way that those aspects can be added to the model in the future.

In the following, Section 2 presents the related works, with a particular focus on BFT and GCM components. Then, we describe our fault-tolerant application and its modelling in Section 3. Finally, Section 4 describes the distributed model checking phase and the properties we verify.

2 Background and Related Works

2.1 Formal Methods for Component Models

As the formal methods matured, they have been integrated into environments that support the development of component-based systems. They ensure the correct behaviour of the assembly of complex applications in all the stages of the development lifecycle (from specification to execution). However, although those frameworks share the same basic concepts, they substantially differ in the range of application domains and supported features. For instance, some of them
are dedicated to embedded systems verification [10, 4] while the others are dedicated to software engineering. We focus below on related works for behavioural specification and verification of distributed components.

Creol [19] is a programming model featuring active objects, requests and futures, similarly to our approach. A framework provides component modelling for Creol; it provides a formal language [13] that supports compositional reasoning and makes automatic testing and verification possible. This language is defined over communication labels, and specifies components in terms of traces of observable behaviour at the interfaces.

Cadena [16] is an environment for modelling and verifying CCM component-based systems. The framework offers a rigorous type-based language [20] for describing component connectors, and the interaction between them. The compositional analysis is based on the assume-guarantee reasoning. However, the component model does not support hierarchical structure.

SOFA [24] is a framework for developing distributed systems. It supports component-based development as well as formal verification. The SOFA 2 component model is hierarchical and supports reconfiguration, making it quite close to ProActive/GCM even though one-to-many communication and asynchrony with futures are not offered by default in SOFA. SOFA uses “behaviour protocols” for specifying possible interactions between components and checking the correctness of the assembly, making the verification process in SOFA quite different from ours, but our approach could also be applied to SOFA components.

This article relies on the pNets [1] formalism for describing the behaviour of parametrized networks of LTSs. We showed in [1] how to build models for GCM components, asynchronous communication, and futures. [7] describes how to specify group communication in pNets. Additionally to faulty components, this article extends the preceding semantics by specifying one-to-many communication at the GCM level, and the management of proxy instances.

The CADP toolset [11] is one of the prominent platforms for the specification, verification, and testing of distributed systems in the academic landscape. It handles several input formalisms, and provides an extensible API. The toolset includes engines for building hierarchically the state-space of systems, building and manipulating LTSs on distributed infrastructures, minimizing LTSs along several behavioural equivalences, model-checking properties, checking equivalences between systems, building test suites, evaluating performances, etc.

2.2 Verifying Byzantine Fault-tolerant Systems

Byzantine fault tolerance (BFT) has a long history [22, 26]; results in this research area are very difficult to obtain and to prove. Indeed, BFT supposes that a faulty process can have any behaviour. The name BFT comes from the original problem raised by Lamport relying on Byzantine generals that must all take the same decision (attack or retreat), knowing that some of the generals are traitors. Traitors can say anything to the others, but the others must all act identically.

In computer science, this situation represents either a faulty process behaving
“randomly” or a malicious entity. BFT has gained new interests since the appari-
tion of a new form of large scale distributed computations relying on entities that, by nature, cannot be trusted. Typically a P2P storage application cannot make any assumption on the kind of misbehaviour the peers can have.

The purpose of this paper is not to prove that a BFT protocol is correct but to understand whether it is possible to represent all the aspects of a complete component application communicating by request-replies, and at the same time reason about the fault-tolerance of this entire application. We focus on a specific application similar to [21] but simplify it: our application consists of a Master component replicating data to be stored on several workers. The master updates the worker value, and gathers replies from workers to retrieve the stored value. If enough non-faulty workers are instantiated, and enough identical replies are returned to the master, the stored value can be retrieved. The objective of this paper is not to study the implementation of the component model, this is why we make the assumption that communications are performed safely. More precisely, we suppose that the middleware ensures that messages systematically follow the bindings, and that a component can only reply to the requests it received. For example, a faulty component cannot communicate to any component of the application, and a faulty component cannot reply instead of a non-faulty one.

Note that the master is supposed to be non-faulty; Protocols for dealing with a faulty master exist and have been heavily studied and implemented. For example, recently [21] implemented a BFT storage in the same settings as our application. Here we simplify the problem and focus on the correct handling of faulty workers, similarly to the case studied in Section 4.2 in [26]. If \( f \) is the number of tolerated faults, \( 2f + 1 \) slaves are sufficient for reaching a consensus. However, as it is generally required in BFT, i.e. when the master can be faulty, we instantiate \( 3f + 1 \) slaves. Section 4 will show that specifying a whole application with those simplifying hypotheses already requires the full power of distributed model-checking over a cloud-like architecture.

Our approach for encoding Byzantine faults is the following: faulty slaves can feature any behaviour, upon verification the model-checker will then explore all the possible behaviours, including the malicious ones. We then specify a simple agreement procedure where the Master component waits until enough slaves answered correctly. In order to count them, our architecture description is aware of which slave is faulty, but the business code does not use this information.

2.3 Distributed Components and their Semantics

This section recalls the component structure and semantics of GCM, a complete definition can be found in [17].

Component Structure. The structure of GCM components is inherited from Fractal: A GCM component can be either composite (i.e. composed of subcomponents), or primitive (a basic element encapsulating the business code). A component comprises a content (providing the functional code) and a membrane (a container managing non-functional operations). The interfaces are the only ac-
cess points to components. Each interface is either client (emitting invocations) or server (receiving invocations). We distinguish functional interfaces addressing the business of the application from non-functional ones invoked to manage, monitor, and introspect the application. A binding connects a client interface to a server interface (Fig. 2); a message emitted by a client interface is transmitted to the server interface bound to it. In composite components, interfaces are either internal – exposed to the subcomponents – or external – exposed to other components. The interface cardinality indicates how many bindings can be made from or to this interface. In this paper, we only use two interface cardinalities: singleton (one-to-one binding) and multicast (one-to-many binding). The different parts of a GCM component are shown in Fig. 1, whereas Fig. 2 shows an assembly of components bound together, on the left there is a composite composed of two primitives; the figure also illustrates different bindings.

**Communication.** The basic communication paradigm in GCM is asynchronous message sending; communication consists in synchronously dropping a message in a request queue at the receiver side, and creating a future to represent the result of the invocation. A future is an empty object representing the result of a computation performed in parallel. Once the future is created, the execution continues immediately on the sender side. When the request treatment is finished, the result is automatically returned to replace all the references to the corresponding future. When a component accesses a future, it is blocked until the result is returned. However, future references can safely be passed between components, inside invocation parameters, or inside a request result. To prevent shared memory between components, parameters and results are copied; no object is passed by reference.

A multicast interface is a client interface that transforms a single invocation into a list of invocations, sent in parallel to a set of connected interfaces. The result of an invocation on a multicast interface is a list of results. Invocation parameters can be distributed according to a distribution policy that can be customized. Typical distribution policies include broadcast that sends the same parameter to each connected component, and scatter that splits the parameter.

**Component Behaviour.** Primitive components encapsulate the business code, their behaviour is highly dependent on the application; it is provided by the
application programmer. The only constraints they must respect are: they serve requests of the request queue, they emit new requests on their client interfaces, and can receive a result for the futures they hold. We consider here only mono-threaded primitive components: a single request is served at a time.

By contrast, composite components have a predefined behaviour: they serve requests in the reception order, and delegate the requests to sub-components, according to the bindings. For example, when a composite component receives a request from the outside, it delegates its service to one of the sub-components.

3 Our Fault-tolerant Application and its Specification

This section describes informally our application, and then presents its behavioural model. We present the architecture using the pNets model [1], a formalism to encode labeled transition systems with value passing, parametrized topologies of processes, and different types of communication. We describe then the primitive component internal behaviour, and the semantic-level process generated from the GCM architecture. We focus on the parts of the specification that are directly related to one-to-many communications and fault-tolerance, details of the other processes are given in [6].

3.1 Distributed Component for Fault-tolerant Storage

Fig. 3 shows the architecture of our application. It consists of a main composite component BFT-Composite. The white part of the composite is the functional content made of a Master component and several slaves. Some of those slaves are called good slaves, i.e. non-faulty, the bad ones are faulty and behave randomly. In practice one never knows which of the slaves is good or bad but it is necessary that the verification process knows this information to be able to count the number of good and bad slaves.

Properties of Interest. From a high-level point of view, we are interested in the storage properties of our application: the stored value can be retrieved unchanged, even if some of the slaves are faulty. Of course, some additional properties are crucial like: the master always finally answers to the requests it
receives. Also, the master must rely on the slaves for storing the value, and does not distinguish good slaves from bad slaves, for example, for writing data the master must broadcast a write request to all the slaves.

3.2 Architecture

We describe here the architecture of the semantic model of our use-case. The overall architecture of the system is shown in Fig. 4. It is composed of:

- An indexed family of slaves receiving invocations from the master. Each of them has a queue storing the requests not treated yet, a body part describing how to treat the incoming requests and delegate them to the behavioural specification of methods Write, Commit, and Read. Each requests can reply to the master by updating a future (represented by the arrows between the Write box and the CO element). The system is instantiated with 3 good slaves and 1 bad slave.

- A Master component receiving requests from a client and forwarding them to the slaves (that are bound to it). It also has a request queue and a body delegating the treatment of requests to sub-parts of the master. Treatment of read and write methods will be detailed below.

- The connections that are one-to-one bindings, except for BC (broadcast) that dispatches a request from the master to all the slaves it is bound to, and CO (collect) that carries a reply from one of the slaves to the appropriate proxy. Those 2 bindings will be detailed in Section 3.3.

To optimize the size of the model, the composite has no request queue and calls are directly issued to the Master component. This has no consequence

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4 We generate the behaviour of each request queue as an individual process able to store a finite number of requests with their parameters.
because the requests are directly delegated to the Master component, and the request queue of the Master is sufficient for dealing with asynchrony.

3.3 The Master Body and its methods

Let us first describe the communication patterns and name conventions that we use in this paper. All local methods are triggered by a first outgoing communication of the form \texttt{!Method}, then the response is received as parameter of a \texttt{?R.Method} incoming communication. For example, in Fig. 6 \texttt{!Get.Write.Proxy} requires a new group proxy for invoking the Write method on the slaves. The proxy is returned and stored into \texttt{p1} by the reply: \texttt{?R.Get.Write.Proxy(p1)}. On the other side, method invocation towards remote components are of the form \texttt{!Call.Method}, those method invocations enqueue a request in the remote request queue, and pass a proxy reference as one of the parameters of the invocation. The remote method will, upon termination, fill the proxy with the calculated value; for this, the \texttt{!R.Method} transition synchronizes at the same time with the invoker that receives the value and with the body of the component containing the method, so that next request can be served.

The master body. The body is encoded in generic way: it serves sequentially functional and non-functional requests. In this work, we only use the service of each functional request (on method Read, Write, or SetF). This service calls the adequate method (e.g., \texttt{!Call_Read}), and waits until the method terminates, signaled by \texttt{R.Method} events (e.g., \texttt{?R_Read}); \texttt{R.Read} synchronizes both with the component that triggered the request and with the body. As requests are served one after the other, this encodes a mono-threaded behaviour for the master.

![Diagram](image-url)

Fig. 5. Behaviour of the method: MasterCollateReplies
**The Attribute controller.** In Fractal, the attribute controller provides read and write access to the attributes of the components; the only attribute of the Master component is $f$ – the number of faults that can be handled. The behaviour of the attribute controller is very simple: it simply provides a setter ($ACSet$) and a getter ($ACGet$) method for storing and retrieving the value of $f$.

**The Collate method.** Based on the vector of replies received by the proxy, this method computes a consensus in order to know whether enough slaves returned a correct answer. It is used by the methods Read and Write described below. Fig. 5 represents the behaviour of Collate in a format similar to Statecharts [15]; starting from initial state $S_0$, Collate is always used by first triggering a ?CollateReplies sending it a vector of replies currently known; then from state $S_1$, a complex transition counts the number of True and False in the vector. It stores in agreed_bit the reply the most frequent and returns (by !R_CollateReplies) the number of replies that agreed on this value. Then, the agreed value can be retrieved by a ?GetBit, that returns the agreed_bit value.

**The Write method.** The write method is the most complex method of our example, it is shown in Fig. 6. It first gets the current value of $f$, read from the

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**Fig. 6.** Behaviour of the Write method
attribute controller, and initializes the variables agree, awaited, and nb_Slave. It consists of two phases; first, a write request is sent to all the slaves, then the master waits until enough slaves agree on the reply, agree is the number of necessary identical replies, and awaited is the number of awaited replies. If necessary, additional replies are awaited, and awaited is incremented. It is not possible to wait for more replies than the number of slaves; if such a situation occurs, it means that the BFT hypothesis is not verified, more exactly, more than f slaves are faulty and an error is raised. When enough identical replies have been received, the write method enters a commit phase that behaves similarly to the write phase. At the end the method returns to the initial phase, emitting a \texttt{!R\_Write} that also indicates the end of the method.

**The Read method.** The behaviour of the Read method is very similar to the Write method above. The main difference is that, after triggering remote invocations and waiting for enough identical replies, it inputs the agreed bit found by the collate method and returns this value to the client.

**The Master Proxies**

**Managing groups of slaves.** We first focus on the management of groups of slaves, i.e. groups to which the write, read and commit requests will be addressed. The part of the pNets that deal with this aspect is shown in Fig. 7. It includes a proxy manager (Fig. 8) that returns an available proxy through its \texttt{Get\_Proxy} invocations. If reconfiguration was enabled, it would receive bind and unbind requests for adding or removing slaves. When a new proxy is requested, one proxy is activated (among the families of \texttt{Proxy\_write}, \texttt{Proxy\_Read}, or \texttt{Proxy\_Commit} proxies), and given the group \(g\) on which next invocation will be performed. A reference to this proxy is returned, and can be used to remotely invoke Write,
Read, or Commit on the slaves. The group $g$ passed upon activation is used later inside the broadcast communication: the circle $BC$: $Q.Write(\ldots)$ performs a synchronization involving the proxy and all the slaves of $g$ sending them the same invocation, $!Slave[i].Q.Write(p,b)$, where $p$ is the proxy identifier. The symmetric communication is performed by the $CO$: $R.Write(p)$ that collects replies from all the slaves of $g$ and returns them to the $Proxy.Write pNet$: each member of $g$ can send a reply to the master. Note that $g$ can be modified inside the manager and a copy of the group is passed upon activation of a proxy. This guarantees that the $CO$ operation will be performed on the same group as the invocation, even if, in the manager, the group is changed in the meantime.

**The Write proxy.** (see Fig. 9) Upon activation, the write proxy waits for an invocation from the master write method. It then initializes the $wRep$ array of received replies as well as $len$ – the number of replies currently received. Its
two main behaviours are then (1) to receive a reply from an element of the group, which updates the \texttt{Wrep} array, and the \texttt{len} value; and (2) to fulfill a \texttt{WaitWrite} invocation from the master write, which returns the current array of received replies once the number of awaited replies is reached. Proxies for read and commit method are similar to the write request proxy.

### 3.4 The Slave Components and their methods

The behaviour of the slaves is much simpler than the one of the master. We encode two kinds of slaves: good slaves behave as expected, whether bad slaves behave randomly and encode the byzantine faulty processes. We instantiate as many faulty processes as the number of faults we can tolerate. The fact that the system description distinguishes between faulty and non-faulty processes has no influence here because the functional parts of the components never use this knowledge: the code of the Master component never distinguishes between the communications towards the faulty slaves, and towards the non-faulty ones.

The slave body serves successively the requests (Commit, Read and Write) arriving at the slave queue much similarly to the master body. The bad slaves and the good slaves have the same body; they all serve the request in a FIFO order, and no two requests are served at the same time: the slaves are mono-threaded. The slaves have three methods: Write, Read and Commit; we show the method \texttt{Write} for the good and bad slaves in Fig. 10, the behaviour of a good slave consists in storing the bit value \( b \) received thanks to a call to \texttt{!SetBit} that sets a local attribute of the slave. There is a method \texttt{!GetBit} for reading this value, it is called by the \texttt{Read} request. The bad slave as shown in Fig. 10 replies randomly to each individual request. The commit phase is here to show how a commit phase would be implemented, but it is not used by our slaves: it would be useful if the master could also have a faulty behaviour.

According to the BFT hypothesis, a bad slave can behave arbitrarily. However, we have to restrict a little this behaviour so that it can be encoded and verified by finite model-checking techniques. Here are the hypotheses we make and the reasons why it is safe to make them:

- Bad slaves do not steal the identity of another entity: we suppose here that the underlying middleware guarantees the identity of the components sending requests or replies. It is the classical “oral messages” assumption of [22].
Bad slaves only reply to required requests. We suppose again that the middleware verifies this to guarantee the integrity of the program execution.

Bad slaves only reply to requests in the order required. This assumption is stronger but we can show that it has no influence on the final result. First, the master is single threaded, waits for enough replies before requiring another computation, and does not access the future afterward; thus late replies would have no influence on the computation. In principle, a bad slave could serve the request in the wrong order and use this information to behave in a malicious manner; but the exhaustive exploration of all the possible replies is even more general than the scenarios using out of order service of requests.

4 Building the model, and running the verification tools

In this section we describe the methods and tools used to build the behavioural model of our application and to check its properties, and we discuss the combination of advanced techniques we have used to master the model complexity.

We build the behavioural model of our case-study in three steps (Fig. 11). From the specification of the component architecture and behaviour, our tool ADL2N [8] builds a hierarchical and parametrized pNets model, including the data types, the behaviour, and the architecture of the system. Then abstractions are applied on the data domains, yielding a finitary model. Finally the model is encoded using a combination of several input formalisms from the CADP toolset [11]: the Fiacre language [5] provides syntax for data types and expressions, definition of LTS, and a form of composition of processes by synchronization on channels; the EXP and SVL languages [11] support the hierarchical encoding of our pNets, and the scripting of the various verification tasks.

Then we run a combination of CADP tools, the most important ones are: cesar.open for generating transition systems from Fiacre programs, either on a single machine, or on parallel infrastructures when used in combination with distributor; exp.open to build product of transition systems described in EXP format; and Evaluator4, the new version of the model-checker that deals with the MCL (Model Checking Language) logics [23], which is an extension of the alternation-free regular \( \mu \)-calculus with facilities for manipulating data.

The Vercors\(^5\) tool platform should assist the programmer in the encoding and verification of his application. It includes the Vercors editors, the ADL2N, ABS and N2F tools; it is currently under development. For this paper, we already have been able to generate approximately 50\% of the Fiacre and EXP code.

One goal of this work is to experiment with various methods for mastering the state explosion inherent to large models, such methods consist of:

1. data abstraction
2. hierarchical hiding and minimization
3. use of contextual environment information
4. distributed state-space generation

\(^5\) http://www-sop.inria.fr/oasis/index.php?page=vercors
We have used 1) in several ways. First, all data variables have been given abstract types with (very small) finite domains, in fact we choose the smallest abstract domain that preserves the formulas to be proven. Secondly, the topology parameters of the system (the number of slaves and number of proxy instances) have been reduced to a minimum number, though significant for our scenario; proving properties that would be valid for any values of such parameters is out of the scope of model-checking. Finally, the request queues raises another issue: their explicit representation has a size exponential in the number of values that the queue cells admit. Our approach is to encode a (small) finite model of the queue, including events denoting an error when this finite queue is out-of-bounds. Then we check by model-checking whether this event is reachable, or the chosen size is sufficient. The soundness of these approaches is worth discussing; for the domains of value-passing parameters, we can define finite abstractions that preserve safety and liveness properties [9]; for the length of queues, we are building an under-approximation, and we check explicitly its validity. But for topology parameters, we have no such general result and we only prove properties for a given instantiation, that is already very helpful as a “debugging tool”. Proving more general properties is not in the scope of this paper.

Method 2) is now quite classical when using bisimulation-based tools. Let us remark that to be optimal, we have to generate models specifically for each formula to prove. Method 3) has been proposed and advocated by the CADP developers, and is indeed very important when combined with 2). The problem arises when you build subsystems hierarchically without taking into account the specific way in which other pieces of the system interact with a given subsystem. The context information can be built automatically by the CADP tools from the behaviour of the other subsystems (in which case it is guaranteed to be sound), or can be specified manually (that may lead to under-approximations). We chose the second option, and we used the context behaviour to reduce further the possible values of input data of some methods, by symmetry arguments.

Method 4) is a hot research topic. We are using a local Cloud platform, providing large computing resources (>1300 cores and 3 Tbytes of RAM), where we can submit jobs in the form of task workflows. In our case, tasks consist of compilation of input formalisms, generation of transition systems for subsystems,
minimization and product of systems, and model-checking. Tasks can be parallel,
but for the current version of CADP, only LTS generation can run in a distributed
way [12]. We were able to build systems with more than $10^9$ states explicitly
stored in distributed memory [18], but then the bottleneck is the merging of this
structure before minimization or model-checking on a single machine. In practice,
the good strategy is to decompose the system in such a way that subsystems
are of reasonable size, or can be strongly constrained by contextual information,
and to run concurrently the tasks computing the behaviour of each subsystem.
Then minimization, product, and model-checking tasks are run as soon as their
inputs are available, in a coarse-grain concurrent workflow.

**Parameter Domains and System Sizes.** We ran the use-case with 3 good
slaves and 1 bad slave, allowing for 1 failure. We also generated the model in
two different configurations, with the length $q$ of the Master Queue respectively
2 (for OutOfBounds detection) and 1 (for optimization).

As we do not have yet enough tool support at the level of the formalism
compilers, we had to do a significant part of the Fiacre/Exp/SVL programming
by hand, so we chose to build one single model with enough events visible to prove
our formulas of interest. The intermediate code consists in 43 Fiacre processes
for a total of 2900 lines of code, and of 330 lines of synchronization vectors in
EXP format encoding 240 pNet structures.

Then the system is divided in 12 subsystems (9 for the Master itself); each
part is encoded in a Fiacre source file, and its state space computed using distri-
butor. So we have at this level 12 independent tasks in our workflow, running
on 2 to 10 cloud nodes each. Each resulting automaton is reduced by branch-
ing bisimulation (with as much local actions hidden as possible), before being
composed in a hierarchical way, using 4 synchronization products. The final
product is minimized again, before running Evaluator4 for checking our proper-
ties. Decomposing the system in an efficiently manner currently requires human
operation: the choice of subsystems is a compromise between: identifying pro-
cesses that may be reused easily (through relabeling); defining subsystems that
are big enough to take advantage of a distributed generation; choosing pieces
which environment behaviour is well-specified.

The system sizes (states/transitions, after minimization) and computation
times are summarized in the following table:

<table>
<thead>
<tr>
<th>Q. size</th>
<th>Queue</th>
<th>Intermediate</th>
<th>Master</th>
<th>GoodSlave</th>
<th>Global</th>
<th>Total time</th>
</tr>
</thead>
<tbody>
<tr>
<td>Q=1</td>
<td>21/229</td>
<td>542/3107</td>
<td>2M/45M</td>
<td>744/6550</td>
<td>22K/110K</td>
<td>10'</td>
</tr>
<tr>
<td>Q=2</td>
<td>237/3189</td>
<td>542/3107</td>
<td>5.8M/103M</td>
<td>5936/61K</td>
<td>34K/164K</td>
<td>59'</td>
</tr>
</tbody>
</table>

The middle columns in the table give reduced sizes for the most interesting sub-
systems: the Master queue, the biggest intermediate subsystem in our decom-
position of the Master, the whole Master component, the (good) Slave component,
and finally the global system, comprising the Master, 4 Slaves, and a Client. The
last column gives the global computation time.

**Correctness Properties.** Once the behavioural model generated, we verified
several properties, written using the MCL logics; they express various facets
of the system correctness. Some properties express global correctness of the application, seen from the (external) client point of view. Others require the visibility of some internal events of the system, and reveal the feasibility of several scenarios, or the impossibility of some errors.

Let us start with simple reachability properties: all requests (Write or Read) sent to the system can terminate and return successfully. The first formula means that for each possible value of fid (the identifier of a client request), the action $R_{Read}$ denoting the return of the corresponding Read request is reachable with some returned value val. This property is True, meaning that the Read request can terminate (this holds also for Write requests).

$$\forall fid: \text{nat} \text{ among } \{0...2\}. \exists b: \text{bool}. \langle \text{true}^* . \langle R_{Read} !fid !b \rangle \rangle \text{ true}$$

Next formula checks the reachability of the BFT Error events. This property is False, meaning that we instantiated enough good slaves.

$$\langle \text{true}^* . \langle \text{Error} (\text{NotBFT}) \rangle \rangle \text{ true}$$

We then ensure that the Master’s queue cannot receive too many requests. Its validity depends on the system client(s). Here we have proved that a queue depth of 1 is sufficient to prove all of our correctness properties, if we have a single client, and if this client waits for replies before sending the next request.

$$\langle \text{true}^* . \langle \text{Error} (\text{Master-OutOfBounds}) \rangle \rangle \text{ true}$$

Also, we have proved Inevitability properties like the following one. It ensures that it is (fairly) inevitable that after a Write request, either the system sends the corresponding Write response or raises an error. Here fairness means "fair reachability of predicates" in the sense of Queille and Sifakis [25]:

$$[ \text{true}^* . \langle (Q_{Write} ?fid: \text{natt} ?bit: \text{bool}) \rangle . \langle \text{not} (\langle \text{Error}.* \rangle \text{ or } \langle R_{Write} !fid \rangle) \rangle \rangle ]$$

$$\langle \text{not} (\langle \text{Error}.* \rangle \text{ or } \langle R_{Write} !fid \rangle) \rangle \rangle \text{ true}$$

Similarly, we have shown that it is fairly inevitable that Read requests are replied, and also that the system is functionally correct: after a Write request (and before the next one), a Read request will answer with the correct value.

To summarise, we proved by model-checking that our application consisting of 1 master and 4 slaves (3 good ones and bad one) behaves correctly: 1) it answers to Read and Write requests, 2) the answers are correct in the sense that the read value is the value that has been written, 3) for this it relies on the slaves for storing the data (the master only performs a consensus), and 4) enough good slaves have been instantiated and the NotBFT error cannot be raised.

5 Conclusion

This paper shows the modelisation and verification by model-checking of a system that features: one-to-many communication, asynchronous communication
with futures, byzantine faults, replication, and consensus. We showed here the possibility to encode and verify the correct behaviour of a whole distributed application that tolerates some faulty processes. Handling byzantine faults is a difficult task, because no assumption can be made on the behaviour of the faulty processes. Such a random behaviour makes automatic verification of the correction of a whole application even more difficult because a lot of possible states must be considered.

A next step could be to integrate the generation of faulty process, replication management, and consensus methods to our specification environment: the user would identify the possibly faulty components and the environment would generate BFT-like behaviour and replication for those components, but also broadcast and consensus operations. The new system could then be model-checked to decide whether the whole application is fault-tolerant.

Another lesson drawn here is that the behaviour of the whole application is huge, we used all the power of the distributed version of CADP on a cloud-like environment to verify the application. This shows that application-level fault-tolerance can be verified by a model-checker, but also that adding any other feature to the system (e.g. reconfiguration for changing the number of replicates at runtime) may be very difficult. To master such complexity we should use semantic properties of the programming model and of the middleware to get better and smaller abstractions at the level of the generated behaviour.

References


Reducing the Model Checking Cost of Product Lines Using Static Analysis Techniques

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Abstract. Software product line engineering is a paradigm to develop software applications using platforms and mass customization. Component based approaches play an important role in development of product lines: Components represent features, and different component combinations lead to different products. The number of combinations is exponential in the number of features, which makes the cost of product line model checking high. In this paper, we propose two techniques to reduce the number of component combinations that have to be verified. The first technique is using the static slicing approach to eliminate the features that do not affect the property. The second technique is analyzing the property and extracting sufficient conditions of property satisfaction/violation, to identify products that satisfy or violate the property without model checking. We apply these techniques on a vending machine case study to show the applicability and effectiveness of our approach. The results show that the number of generated states and time of model checking is reduced significantly using the proposed reduction techniques.

1 Introduction

Software product line engineering is a paradigm to develop software applications using platforms and mass customization. To this end, the commonalities and differences of the applications should be modeled explicitly [1]. Feature models are widely used to model the variability of software product lines. A feature model is a tree of features, containing mandatory and optional features as well as a number of constraints among them. A product is then defined by a combination of features, and product family is the set containing all of the valid feature combinations [2]. A configuration vector can be used to keep track of inclusion or exclusion of features.

The Vending Machine Example: Feature Model. Throughout this paper, we use a product family of vending machines as a running example. A vending machine may serve coffee and/or tea. It also may add milk to the coffee. Figure 1 shows the feature model of the family of vending machines.

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Software product line engineering enables proactive reuse by developing a family of related products. One of the main approaches to develop software product lines is the compositional approach, in which features are implemented as distinct code units [3]. These code units are reused when the corresponding units are composed to generate each product. Component technology [4] is suitable in this approach as reusability is an important characteristic of software components. In component-based development of product lines, each feature is implemented using a component. Some of the features can be implemented within the components in a fine-grained manner as well, using annotative techniques [5]. Consequently, the behavior of a component may change according to inclusion or exclusion of the features. Software product line engineering is used in the development of embedded and critical systems [6]. Therefore formal modeling and verification of software product lines is essential.

Model checking [7] is a promising technique for developing more reliable systems. Recently, several approaches have been developed for formal modeling of product lines [8–13]. These approaches capture the behavior of the entire product family in a single model by including the variability information in it. In other words, it is specified in the model how the behavior changes when a feature is included or excluded. Model checking of product lines is discussed in [10, 12, 13]. In these approaches, the model checker investigates all of the possible feature combinations when verifying the model of a product family against a property, and the result of model checking is the set of products that satisfy the given property. The focus of these works is on adapting model checking algorithms to verify product families, and they do not address the state space explosion issue. However, the main problem of model checking is its high computational and memory costs which may lead to state space explosion. This problem limits the applicability of model checking technique to verify product lines, as in product families the number of products can be exponential in the number of features. In [14, 15], two incremental approaches are proposed for product line verification. In [14], only sequential composition of features is discussed which is a considerable limitation as the approach is not applicable to concurrent systems. The focus of [15] is on reducing the effort of applying deductive verification techniques (not model checking) on product lines. The main idea of our approach is to use static slicing and static analysis techniques to tackle the state space explosion problem in model checking of component-based software product lines.

We use Rebeca to model product families in a component-based manner, as a basis to explain our approach. However, the approach is not limited to Re-
beca models, and it is applicable to any modeling language with slicing analysis support. In our approach, each feature is modeled using one component that captures its corresponding behavior, or using an alternative behavior within a component that changes the behavior of the component based on the presence or absence of the feature accordingly. Each product contains the components associated to the features that are included in the product, and the behavior of each of its components is determined according to the features that are included/excluded in that product. The model checker considers all of the possible combinations of components and alternative behaviors, to verify the product family. The focus of this paper is on reducing the number of combinations that should be investigated in model checking. We propose two techniques for this purpose.

The first technique uses the static slicing approach. Static slicing [16] is an analysis technique that extracts the statements from a program that are relevant to a particular computation. This technique has been used as a reduction technique in model checking of Promela [17], CSP [18], Petri-nets [19], and Rebeca [20, 21] models. In [22], an evaluation of applying static slicing for model reduction is presented. The result shows significant reductions that are orthogonal to a number of other reduction techniques, and applying slicing is always recommended because of its automation and low computational costs. One of the main approaches for slicing is using reachability analysis on program dependence graph. The nodes of a program dependence graph are the statements of the program, and its edges represent data and control dependencies among the statements. In this paper, we adapt the program dependence graph and the reachability algorithm, to use static slicing to identify the features that do not affect the correctness of the property. By discarding these features, the model checker investigates fewer feature combinations when model checking the product family.

In the second technique, we analyze the property statically to extract sufficient conditions of its satisfaction or violation. These conditions are used along with reachability conditions for variables to conclude satisfaction or violation of the given property for certain products, without verification. The model checker does not verify these products, therefore the number of feature combinations that should be verified is reduced. It should be noted that the proposed techniques (slicing, extracting conditions from property, and investigating reachability of variables) can be applied automatically.

This paper is structured as follows. Section 2 explains how product families are modeled and model checked. In Section 3 we describe the slicing technique that is used to identify the features that do not affect a property. Section 4 describes our approach for extracting sufficient conditions of property satisfaction/violation, and identifying products that satisfy or violate the property, without model checking. In Section 5 we present the results of using the two proposed techniques for reducing the feature combinations of a vending machine case study. Finally, we conclude our work in Section 6.
2 Modeling and Model Checking Product Families

This section introduces the Rebeca modeling language [23], and explains how a product family can be modeled and model checked using Rebeca. We select Rebeca as a basis to describe our approach, because it is suitable for modeling concurrent systems, it is supported by the Modere model checking tool [24], it supports components [25], and the slicing technique is adapted to be applicable on Rebeca models [20, 21]. However, our proposed approach is not limited to Rebeca models, and can be applied to other modeling languages with similar facilities as well.

2.1 Rebeca

Rebeca is an actor-based language for modeling concurrent and distributed systems as a set of reactive objects which communicate via asynchronous message passing. A Rebeca model consists of a set of reactive classes. Each reactive class contains a set of state variables and a set of message servers. Message servers execute atomically, and process the receiving messages. The initial message server is used for initialization of state variables. A Rebeca model has a main part, where a fixed number of objects are instantiated from the reactive classes and execute concurrently. We refer to these objects as rebecs. The rebecs have no shared variable, and each rebec has a single thread of execution that is triggered by reading messages from an unbounded message queue. When a message is taken from the queue, its corresponding message server is invoked. In [25], components are added to the Rebeca language to encapsulate tightly coupled reactive objects. In other words, a component is a set of one or more reactive objects.

2.2 Product Family Model

To model product families, we should model optional components (which may be included in some of the products, and excluded in other products), and alternative behaviors of components. Different combinations of optional components and alternative behaviors lead to different products. To this end, we use a special tag \( @AC \) before a statement to specify the application condition of the statement. An application condition is a propositional logic formula in terms of features. This tag indicates that the statement will be executed only in those products that \( AC \) holds. When a feature \( F \) corresponds to a component, we use \( @F \) tag before all the message server calls to that component. Subsequently, message servers of a component are invoked only if its associated feature is included in a product. If the feature is excluded in a product, no message is sent to its corresponding component, and the component will be excluded. Moreover, these tags can be used to indicate the change of the behavior within components according to presence and absence of features.

The Vending Machine Example: Rebeca Model. Figure 2 shows the Rebeca code for the product family of vending machines. In this model, there is a
controller component that manages coffee and tea requests and sends messages
to the coffee maker and tea maker components accordingly. The nextRequest
message server (line 12”) is responsible for handling the requests. When there
is request for coffee (req = 1), the serveCoffee message is put in the queue of
coffeeMaker; if the machine is capable of serving coffee (line 15”). If the machine
does not have the coffee option, the coffee request is ignored and the machine
processes the next request (line 17”). The tea request (req = 2) is handled in a
similar way. Consequently, if the coffee or tea feature is excluded in a product, no
message is sent to the corresponding component, and the component will be also
excluded. In the coffee maker component, the behavior changes according to the
existence of the milk feature. If the milk feature is included in a product, milk is
added to coffee (line 15). One of the linear temporal logic (LTL) [26] properties
that can be considered for this model is \( P : \square(\neg(addingCoffee \land addingTea)) \),
where \( \square \) stands for globally. This property describes that the machine should
not add both coffee and tea to a drink at the same time.

| 1 reactiveclass CoffeeMaker { | 1 reactiveclass TeaMaker { |
| 2 knownrebecs { | 3 knownrebecs { |
| 3 Controller ctrl; | 4 Controller ctrl; |
| 4 } | 5 Controller ctrl; |
| 5 statevars { | 6 CoffeeMaker cm; |
| 6 boolean addingCoffee; | 7 int req; |
| 7 boolean addingMilk; | 8 statevars { |
| 8 } | 9 boolean addingTea; |
| 9 msgsrv initial() { | 10 boolean addingCoffee; |
| 10 addingCoffee = false; | 11 boolean addingTea; |
| 11 addingMilk = false; | 12 int req; |
| 12 } | 13 statevars { |
| 13 msgsrv serveCoffee() { | 14 boolean addingCoffee = true; |
| 14 addingCoffee = true; | 15 @Milk boolean addingMilk = true; |
| 15 self.serveComplete(); | 16 self.serveComplete(); |
| 16 } | 17 } |
| 17 msgsrv serveComplete() { | 18 msgsrv serveTea() { |
| 18 addingMilk = false; | 19 addingTea = true; |
| 19 addingMilk = false; | 20 addingTea = true; |
| 20 ctrl.nextRequest(); | 21 ctrl.nextRequest(); |
| 21 ctrl.nextRequest(); | 22 } |
| 22 } | 23 } |

Fig. 2. The Rebeca code of the product family of vending machines
2.3 Model Checking the Product Family

For a product line with \( n \) features (where each feature corresponds to a component or an alternative behavior of a component), potentially there exist \( 2^n \) products in its corresponding product family. To model check the product family, a configuration vector \( C \in \{I, E, ?\}^n \) (\( I \): Included, \( E \): Excluded, ?: not decided) is used to keep track of inclusion and exclusion decisions that are made for each feature [10]. The validity of configuration vector with respect to the feature model can be checked during model checking by transforming the feature model to a propositional logic formula [27] and using a SAT-solver (like [28]) to investigate its satisfiability. The result of model checking a product family against a property is the set of products (represented through configuration vectors) that satisfy the given property.

The Vending Machine Example: Model Checking. We assume the first, second, and third elements of configuration vector correspond to Coffee, Tea, and Milk features, respectively. The result of model checking the product family of vending machines against the property \( P \) is:

\[
R = \{\langle E, I, E \rangle, \langle I, E, E \rangle, \langle I, I, E \rangle, \langle I, E, I \rangle, \langle I, I, I \rangle\}
\]

Note that the configurations \( \langle E, E, E \rangle \), \( \langle E, E, I \rangle \), and \( \langle E, I, I \rangle \) do not appear in \( R \) as they do not represent valid products, according to the feature model.

3 Slicing the Model of a Product Family

The main purpose of slicing is to extract the statements of a program that are relevant to a particular computation. A backward program slice consists of the statements that potentially affect the values computed by some statement of interest (referred to as a slicing criterion). A common approach for program slicing is applying a graph reachability algorithm on the program dependence graph. In this section, we first describe the program dependence graph of Rebeca models that capture the behavior of a product family, and then present the slicing algorithm that computes the slice of the product family model, followed by a short discussion on model checking the computed slice.

3.1 Program Dependence Graph

A program dependence graph models the data and control dependencies that exist among the statements of a program. In such a graph, the nodes represent the statements of a program, and the edges are dependencies among them. A data dependence edge exists between two statements if one statement assigns a value to a variable and the other statement may read the value of that variable before it is changed by another statement. A control dependence edge exists between two statements if one statement determines whether the other statement is executed.

A special dependence graph named Rebeca Dependence Graph (RDG), is introduced for Rebeca in [20]. In this graph, there is a class node for each
reactive class, and *member dependence* edges connect the class nodes to their message servers. Each message server is modeled by an *entry* node, a set of nodes representing its statements, and data dependence and control dependence edges modeling dependencies within the body of the message server. Sending a message is represented through an *activation* node. In addition, an *activation* edge is used to connect the activation node to the entry node of the corresponding message server. Finally, *intra-rebec dependence* edge represents the dependency between a statement that writes on a state variable in a message server, and a statement which reads the value of that variable in another message server. To adapt the dependence graph for product families, we add a tag to the nodes to specify their application conditions.

**The Vending Machine Example: RDG.** Figure 3 shows the RDG of the vending machine. In this graph the nodes 15, 15”, 17”, 19”, and 21”, are tagged with a feature as their corresponding statements in the Rebeca model are tagged with these features.

### 3.2 Slicing Algorithm

After constructing the program dependence graph, the slice with respect to a property can be computed using a graph reachability algorithm. The slicing criterion consists of the statements that assign values to the variables that appear in the given property. Figure 4 shows the static slicing algorithm that is adapted to extract the features affecting the property as well. To this end, the algorithm traverses the graph backwards (starting from the slicing criterion nodes), and adds the traversed nodes to the slice, and their corresponding features to the relevant features set. In this algorithm, we assume that $\text{Features}(v)$ gives the set of features that appear in the application condition of node $v$. The features in the set $F$ are the components and the alternative behaviors that their presence
or absence affects the correctness of the property. Therefore, the model checker should investigate their different combinations.

The Vending Machine Example: Slicing. The slicing criterion nodes for the property \( P : \Box(\neg(\text{addingCoffee} \land \text{addingTea})) \), are indicated by gray nodes in Figure 3. The slice computed by the slicing algorithm contains all of the nodes except 11, 15, 20, and the feature set is \( F = \{ \text{Coffee}, \text{Tea} \} \).

### 3.3 Model Checking the Slice

The features that do not exist in the set \( F \) represent the components and alternative behaviors that do not affect the property. Therefore, the combinations of these features can be ignored when model checking the slice of a product family. Having a feature model with \( n \) features, there will be at most \( 2^n \) feature combinations (products), in the product family. By excluding \( m \) features that do not affect the property, the number of products to verify is reduced to \( 2^{(n - m)} \). The configuration vector is \( C \in (I, E, ?)^{(n - m)} \), as practically, the value of an element that its associated feature is removed always remains as ".?".

The result of model checking the slice of product family against a property is the set \( R \) containing the configurations that satisfy the given property. However, these configurations are based on the combinations of \( n - m \) features and do not describe identifiable products. As the other \( m \) features do not affect the property, we can combine the configurations in \( R \) with inclusion and exclusion of each of these features, taking constraints of the feature model into account, to achieve the final result. If we have \( r \) configurations such as \( C \in (I, E)^{(n - m)} \) in \( R \), The ultimate result \( R' \) would contain \((r \times 2^m) - u \) configurations in the form \( C \in (I, E)^n \), where \( u \) is the number of feature combinations that are not valid according the feature model.
The Vending Machine Example: Model Checking the Slice. The Milk feature does not affect the property $P$, and does not appear in the slice. This reduces the number of products in the product family from $2^3$ to $2^2$. The result of model checking the slice against $P$ is:

$$R = \{ \langle E, I \rangle, \langle I, E \rangle, \langle I, I \rangle \}$$

In the next step, the milk feature should be combined with each of the above configurations. So it should be included and be excluded in these configurations (that leads to two new configurations per each configuration). The final result is $R'$ that consists of $(3 \times 2^1) - 1$ configurations ($\langle E, I, I \rangle$ is invalid):

$$R' = \{ \langle E, I, E \rangle, \langle I, E, E \rangle, \langle I, E, I \rangle, \langle I, I, E \rangle, \langle I, I, I \rangle \}$$

4 Static Analysis of Property Satisfaction/Violation in Products

In this section, we describe how satisfaction/violation of a property can be inferred for some of the products without model checking. For this purpose, we extract sufficient conditions for property satisfaction/violation in terms of initial values of atomic propositions and the possibility of their change in the model. We assume that a property is described using boolean variables where each variable corresponds to an atomic proposition. Therefore, we can evaluate sufficient conditions using the initial values of the variables and the possibility of their change in different products. The latter is achieved by analyzing the reachability of statements to obtain a condition in terms of presence and absence of features, which describes in which products the value of a variable may change. Using the result of evaluating sufficient conditions, we determine a subset of products that satisfy/violate the property without model checking. In other words, we indicate in which components and in which of their alternative behaviors the value of a variable does not change, and consequently the property is satisfied/violated.

It should be mentioned that this analysis only makes sense for models of product families that capture the behavior of all products. In traditional model checking, the value of a variable changes when the model is executed, and almost always it is not possible to infer satisfaction/violation of a property without model checking.

4.1 Condition Extraction from the Property

In this work, we consider properties expressed in linear temporal logic (LTL) [26]. An LTL formula over the set of $AP$ of atomic propositions is formed according the following grammar:

$$\varphi ::= true \mid false \mid p \mid \neg \varphi \mid \varphi_1 \land \varphi_2 \mid \varphi_1 \lor \varphi_2 \mid \Box \varphi \mid \Diamond \varphi \mid \varphi_1 U \varphi_2$$
In the above grammar, \( p \in AP \), and \( \square, \Diamond \), and U stand for globally, finally, and until operators respectively.

A transition system \( TS \) is a tuple \((S, Act, \rightarrow, I, AP, L)\) where \( S \) is a set of states, \( Act \) is a set of actions, \( \rightarrow \subseteq S \times Act \times S \) is a transition relation, \( I \subseteq S \) is a set of initial states, \( AP \) is a set of atomic propositions, and \( L : S \rightarrow 2^{AP} \) is a labeling function. For simplicity, in this paper we assume a single initial state \( s_0 \) for a transition system. A state \( s \) is reachable from the initial state, \( s_0 \rightarrow^* s \), if there exists a set of actions \( \alpha_i \in Act \) such that \( s_0 \xrightarrow{\alpha_1} s_1 \xrightarrow{\alpha_2} \ldots \xrightarrow{\alpha_n} s \).

Figure 5 shows the proposed rules for extracting sufficient conditions of property satisfaction/violation. These conditions are statically inferable from the initial values of atomic propositions, and also the atomic propositions that do not vary in \( TS \). The notation \( \overline{\bigvee}_{TS}(\varphi) \) means that the LTL formula \( \varphi \) does not vary in \( TS \), because some of the atomic propositions in \( \varphi \) do not change in \( TS \).

Rules 1-8 are trivial. We can infer \( TS \models \square \varphi \) from \( TS \models \varphi \) (Rule 9) if \( \varphi \) does not vary in \( TS \) (\( \overline{\bigvee}_{TS}(\varphi) \)). From \( TS \models \varphi \) we can conclude that \( TS \models \square \varphi \), as \( \varphi \) should hold in all states and otherwise \( \square \varphi \) is violated (Rule 10). Similar justifications can be made for the other rules.

Using these rules, we extract sufficient conditions for property satisfaction or violation. These conditions are propositional logic formulas in terms of initial values of atomic propositions \( (p \in L(s_0)) \) and their variability \( (\overline{\bigvee}_{TS}(p)) \).

**The Vending Machine Example: Extracting Satisfaction/Violation Conditions.** For the property \( P : \square(\neg(\text{addingCoffee} \land \text{addingTea})) \) we can extract sufficient conditions for satisfaction/violation by applying the rules in Figure 5 in the following order (it is assumed that \( p \) is (\emph{addingCoffee} = true), and \( q \) is (\emph{addingTea} = true)):

\[
\begin{align*}
TS \models (\square(\neg(p \land q))) \quad & \text{if} \quad (TS \models (\neg(p \land q)) \land \overline{\bigvee}_{TS}(\neg(p \land q)) \quad \text{Rule}(9) \\
TS \models (\neg(p \land q)) \quad & \text{if} \quad TS \models (p \land q) \quad \text{Rule}(3) \\
TS \not\models (p \land q) \quad & \text{if} \quad (TS \not\models p) \lor (TS \not\models q) \quad \text{Rule}(8) \\
TS \not\models p \quad & \text{if} \quad p \not\in L(s_0) \quad \text{Rule}(2) \\
TS \not\models q \quad & \text{if} \quad q \not\in L(s_0) \quad \text{Rule}(2) \\
\overline{\bigvee}_{TS}(\neg(p \land q)) \quad & \text{if} \quad \overline{\bigvee}_{TS}(p \land q) \quad \text{Rule}(18) \\
\overline{\bigvee}_{TS}(p \land q) \quad & \text{if} \quad \overline{\bigvee}_{TS}(p) \land \overline{\bigvee}_{TS}(q) \quad \text{Rule}(22) \\
\overline{\bigvee}_{TS}(p \land q) \quad & \text{if} \quad (TS \not\models p) \land \overline{\bigvee}_{TS}(p) \quad \text{Rule}(23) \\
\overline{\bigvee}_{TS}(p \land q) \quad & \text{if} \quad (TS \not\models q) \land \overline{\bigvee}_{TS}(q) \quad \text{Rule}(24) \\
\end{align*}
\]

This way, the three extracted sufficient conditions of property satisfaction would be:

\[
\begin{align*}
TS \models P \quad & \text{if} \quad (p \not\in L(s_0) \lor q \not\in L(s_0)) \land (\overline{\bigvee}_{TS}(p) \land \overline{\bigvee}_{TS}(q)) \\
TS \models P \quad & \text{if} \quad (p \not\in L(s_0) \lor q \not\in L(s_0)) \land (p \not\in L(s_0) \land \overline{\bigvee}_{TS}(p)) \\
TS \models P \quad & \text{if} \quad (p \not\in L(s_0) \lor q \not\in L(s_0)) \land (q \not\in L(s_0) \land \overline{\bigvee}_{TS}(q)) \\
\end{align*}
\]
<table>
<thead>
<tr>
<th>Rule</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Rule(1)</td>
<td>( TS \models p ) if ( p \in L(s_0) )</td>
</tr>
<tr>
<td>Rule(2)</td>
<td>( TS \not\models p ) if ( p \notin L(s_0) )</td>
</tr>
<tr>
<td>Rule(3)</td>
<td>( TS \models \neg \varphi ) if ( TS \not\models \varphi )</td>
</tr>
<tr>
<td>Rule(4)</td>
<td>( TS \models \neg \varphi ) if ( TS \models \varphi )</td>
</tr>
<tr>
<td>Rule(5)</td>
<td>( TS \models (\varphi_1 \lor \varphi_2) ) if ( (TS \models \varphi_1) \lor (TS \models \varphi_2) )</td>
</tr>
<tr>
<td>Rule(6)</td>
<td>( TS \not\models (\varphi_1 \lor \varphi_2) ) if ( (TS \not\models \varphi_1) \land (TS \not\models \varphi_2) )</td>
</tr>
<tr>
<td>Rule(7)</td>
<td>( TS \models (\varphi_1 \land \varphi_2) ) if ( (TS \models \varphi_1) \land (TS \models \varphi_2) )</td>
</tr>
<tr>
<td>Rule(8)</td>
<td>( TS \not\models (\varphi_1 \land \varphi_2) ) if ( (TS \not\models \varphi_1) \lor (TS \not\models \varphi_2) )</td>
</tr>
<tr>
<td>Rule(9)</td>
<td>( TS \models \Box \varphi ) if ( (TS \models \varphi) \land \Box_{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(10)</td>
<td>( TS \not\models \Box \varphi ) if ( TS \not\models \varphi )</td>
</tr>
<tr>
<td>Rule(11)</td>
<td>( TS \models \Diamond \varphi ) if ( TS \models \varphi )</td>
</tr>
<tr>
<td>Rule(12)</td>
<td>( TS \not\models \Diamond \varphi ) if ( (TS \not\models \varphi) \land \Diamond_{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(13)</td>
<td>( TS \models (\varphi_1 \lor \varphi_2) ) if ( TS \models \varphi_2 )</td>
</tr>
<tr>
<td>Rule(14)</td>
<td>( TS \not\models (\varphi_1 \lor \varphi_2) ) if ( (TS \not\models \varphi_1) \land (TS \not\models \varphi_2) )</td>
</tr>
<tr>
<td>Rule(15)</td>
<td>( TS \not\models (\varphi_1 \lor \varphi_2) ) if ( (TS \not\models \varphi_2) \land \Diamond_{TS}(\varphi_2) )</td>
</tr>
<tr>
<td>Rule(16)</td>
<td>( \overline{\varphi}_0 ) if ( \not\exists s</td>
</tr>
<tr>
<td>Rule(17)</td>
<td>( \overline{\varphi}_0 ) if ( \not\exists s</td>
</tr>
<tr>
<td>Rule(18)</td>
<td>( \overline{\Box}<em>0 ) if ( \Box</em>{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(19)</td>
<td>( \overline{\Box}<em>0 ) if ( \Box</em>{TS}(\varphi_1) \land \Box_{TS}(\varphi_2) )</td>
</tr>
<tr>
<td>Rule(20)</td>
<td>( \overline{\Diamond}<em>0 ) if ( \Diamond</em>{TS}(\varphi_1) \land \Box_{TS}(\varphi_1) )</td>
</tr>
<tr>
<td>Rule(21)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \models \varphi_1) \land \Box</em>{TS}(\varphi_1) )</td>
</tr>
<tr>
<td>Rule(22)</td>
<td>( \overline{\Diamond}<em>0 ) if ( \Box</em>{TS}(\varphi_1) \land \Box_{TS}(\varphi_2) )</td>
</tr>
<tr>
<td>Rule(23)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Diamond</em>{TS}(\varphi_1) )</td>
</tr>
<tr>
<td>Rule(24)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Diamond</em>{TS}(\varphi_2) )</td>
</tr>
<tr>
<td>Rule(25)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Box</em>{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(26)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Box</em>{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(27)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Box</em>{TS}(\varphi) )</td>
</tr>
<tr>
<td>Rule(28)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Box</em>{TS}(\varphi_2) )</td>
</tr>
<tr>
<td>Rule(29)</td>
<td>( \overline{\Diamond}<em>0 ) if ( (TS \not\models \varphi_1) \land \Box</em>{TS}(\varphi_2) )</td>
</tr>
</tbody>
</table>

**Fig. 5.** Rules for extracting sufficient conditions of property satisfaction/violation, based on initial values of atomic propositions, and the atomic propositions that do not vary in \( TS \)
A sufficient condition of property violation for $P$ can be extracted in a similar way:

$$TS \not\models P \quad \text{if} \quad p \in L(s_0) \land q \in L(s_0)$$

### 4.2 Evaluation of the Extracted Conditions

The initial values of atomic propositions ($p \in L(s_0)$ or $p \notin L(s_0)$) are computed based on initialization statements. For simplicity, we assume that the property is described using boolean variables only. It should be mentioned that we can always rewrite a property such as $\square(x = y + z)$ in the form $\square(v = \text{true})$, where $v$ is boolean variable representing $x = y + z$. This assumption implies that each atomic proposition is a boolean variable in the Rebeca model, and the value that is assigned to the variable in the initialize message server, determines if $p \in L(s_0)$ or $p \notin L(s_0)$.

The next step is to investigate if the value of the atomic proposition $p$ may vary ($\mathcal{V}_{TS}(p)$). The value of variable $v$ (where $v$ corresponds to $p$) changes in a product if the product has a reachable statement $s$ that assigns a value to $v$. According to our model for product families, a tagged statement is executed when its application condition holds in a product. Other statements are executed normally. We assume that $\mathcal{F}(s)$ gives the application condition that is associated to a tagged statement $s$, and for other ones returns true. A statement $s$ is reachable in a product if its associated application condition holds in the product, as well as at least one of the application conditions assigned to those statements on which $s$ is control/activation dependent (possibly indirectly). We compute the reachability condition of the statement $s$ recursively as:

$$RC(s) = \bigvee_{r \rightarrow_{c,a}s} (\mathcal{F}(s) \land RC(r))$$

In the above computation, $r \rightarrow_{c,a}s$ is the set of statements on which $s$ is control or activation dependent. To avoid recursion, we mark each statement $r$ when its condition is extracted, and in $r \rightarrow_{c,a}s$ we only consider the unmarked statements. Note that when a behavioral model is inconsistent (e.g. $RC(s)$ contains the conjunction of a feature and its negation), the statement $s$ is not reachable in any of the products.

We assume $\text{Def}(v)$ is the set of statements that assign value to the variable $v$, except the initialization statement which is the one assigning value to $v$ in the initial message server of the Rebeca model. The value of $v$ may change in a product, if at least one of the statements $s \in \text{Def}(v)$ are reachable in that product. The atomic proposition $p$ which corresponds to $v$ may vary in transition system $TS$ if:

$$\mathcal{V}_{TS}(p) = \bigvee_{s \in \text{Def}(v)} RC(s)$$

Consequently:
The possibility of variation for $p$ is thus described using application conditions, where each application condition is a propositional logic formula in terms of features itself. Substituting the initial values of atomic propositions and their possibility of variation ($\nabla_{TS}(p)$) in sufficient conditions of property satisfaction/violation, leads to a number of propositional logic formulas. These formulas describe products that we can conclude satisfaction/violation of the given property in them statically. A product satisfies or violates a property if at least one of the sufficient conditions of property satisfaction or violation holds for it, because of the components and alternative behaviors that it includes. The model checker only verifies the products that their satisfaction or violation cannot be concluded from sufficient conditions.

The Vending Machine Example: Evaluation of the Extracted Conditions. We assume that atomic propositions $p$ and $q$ correspond to addingCoffee and addingTea variables, respectively. According to the initializations in the Rebeca model, we can conclude that $p \notin L(s_0)$ and $q \notin L(s_0)$. The statements 10 and 14 assign value to addingCoffee which means that $\text{Def}(\text{addingCoffee}) = \{s_{14}, s_{19}\}$. Therefore:

$$\nabla_{TS}(p) = \neg(RC(s_{14}) \lor RC(s_{19})) = \neg\text{Coffee}$$

Because:

$$RC(s_{14}) = RC(s_{13}) = RC(s_{15}) = \text{Coffee} \land RC(s_{14}) = \text{Coffee} \land RC(s_{12}) = \text{Coffee} \land [RC(s_{17}) \lor RC(s_{15}) \lor RC(s_{16}) \lor RC(s_{17}) \lor RC(s_{19}) \lor RC(s_{21})] = \text{Coffee} \land [RC(s_{17}) \lor RC(s_{16}) \lor RC(s_{20}) \lor RC(s_{21})] = \text{Coffee} \land [RC(s_{17}) \lor RC(s_{16}) \lor RC(s_{20}) \lor \text{true} \lor RC(s_{21})] = \text{Coffee}$$

and:

$$RC(s_{19}) = RC(s_{18}) = RC(s_{16}) = RC(s_{13}) = \text{Coffee}$$

Similarly, we can compute $\nabla_{TS}(q) = \neg\text{Tea}$. By substitution of $\nabla_{TS}(p)$ and $\nabla_{TS}(q)$ with $\neg\text{Coffee}$ and $\neg\text{Tea}$ respectively, the following conditions are achieved which describe the products for which satisfaction/violation of $P$ is inferable without model checking:

$$TS \models P \text{ if } (\neg\text{Coffee} \land \neg\text{Tea})$$

$$TS \models P \text{ if } \neg\text{Coffee}$$

$$TS \models P \text{ if } \neg\text{Tea}$$
According to the above conditions, the products that do not have the Coffee feature, and the products that do not have the Tea feature, satisfy \( P \), and there is no need to verify them. This way, the number of the products that should be model check is reduced to \( 2^2 - 3 \), as we can tell that the products \( \langle I, E \rangle \), \( \langle E, I \rangle \), and \( \langle E, E \rangle \) satisfy \( P \) (although \( \langle E, E \rangle \) is not a valid product).

\begin{figure}[h]
\centering
\includegraphics[width=\textwidth]{feature_model.png}
\caption{The feature model of the vending machine case study}
\end{figure}

\section{Results}

We applied our proposed approach to a vending machine case study that is much more complex than the running example \(^3\). The machine includes a controller that handles the requests. Figure 6 shows the feature model of the vending machine. The coffee maker, tea maker, and soda server components are responsible for serving the associated drinks. There is also a milk adder component which adds milk to coffee. There are two coffee container components and two tea container components, containing black coffee, coffee with cream, black tea, and green tea, respectively. The coffee maker and the tea maker components use the proper container to serve the requested drink. They add water through the water component. The water component can be filled using two different mechanisms which are handled by the filler 1 and filler 2 components. Finally, there are two different payment methods for a vending machine: paying by coin, or paying by card. We defined the following six LTL properties to be verified.

- \( P_1 = \Box[\neg(\text{ServingCoffee} \land \text{ServingTea} \land \text{ServingSoda})] \)
- \( P_2 = \Box[\neg(\text{empty})] \)
- \( P_3 = \Box[\neg(\text{overFlow})] \)
- \( P_4 = \Box[\neg(\text{addingBlackCoffee} \land \text{addingCreamCoffee})] \)
- \( P_5 = \Box[\neg(\text{addingBlackTea} \land \text{addingGreenTea})] \)
- \( P_6 = \Box\Diamond(\text{ServingSoda}) \)

The first property describes that the vending machine should not be serving three drinks at the same time. The second and third properties check that the

\(^3\) The source code is available at http://ece.ut.ac.ir/rrhosravi/sourcecode
water container should not get empty, or overflow. The forth property describes that the machine should not add black coffee together with coffee and cream to a drink. This fact should be also checked for the tea drink (the fifth property). The last property states that the machine should serve soda infinitely often.

Table 1. Number of states and time of verification (in seconds) before applying the techniques (first column), after applying the slicing technique (second column), and after identifying products that satisfy/violate the property without model checking (third column), for the vending machine case study

<table>
<thead>
<tr>
<th></th>
<th>Complete Model</th>
<th>Static Slicing</th>
<th>Slicing and Static Analysis</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>states</td>
<td>time(sec)</td>
<td>states</td>
</tr>
<tr>
<td>P1</td>
<td>-</td>
<td>-</td>
<td>49,307,358</td>
</tr>
<tr>
<td>P2</td>
<td>-</td>
<td>-</td>
<td>39,169,329</td>
</tr>
<tr>
<td>P3</td>
<td>-</td>
<td>-</td>
<td>39,182,632</td>
</tr>
<tr>
<td>P4</td>
<td>-</td>
<td>-</td>
<td>43,484,712</td>
</tr>
<tr>
<td>P5</td>
<td>-</td>
<td>-</td>
<td>47,317,992</td>
</tr>
<tr>
<td>P6</td>
<td>-</td>
<td>-</td>
<td>114,547,805</td>
</tr>
</tbody>
</table>

Table 1 shows the number of states and the time of verification (in seconds) for model checking the product family of vending machine case study. The time of applying slicing technique and computing sufficient conditions are negligible comparing to model checking time and are ignored. The complete model cannot be model checked against the properties because of state space explosion (first column). After applying the slicing technique and eliminating irrelevant features, the sliced model can be checked against the properties (second column). However, the number of states and time of verification can be reduced even more by extracting sufficient conditions of property satisfaction/violation, and identifying products that satisfy/violate the property without model checking.

6 Conclusion

In this paper we presented two techniques to reduce the number of products of a product line that are model checked against a property. This way, the number of generated states and the required time for verifying product families are reduced. The first technique was to apply static slicing to eliminate the features that do not affect the property. The second technique was to analyze the property and reachability of its variables in different products statically to identify products that satisfy/violate the property without model checking. The results of using these techniques in model checking the vending machine case study show the effectiveness of our approach as the number of generated states and time of verification reduced significantly after applying these techniques. The slicing and static analysis technique are completely automatic, and their cost is negligible comparing to the verification cost which makes using our approach for model checking product families practical.
References

Abstract. Archery is a language for behavioural modelling of architectural patterns, supporting hierarchical composition and a type discipline. This paper extends Archery to cope with the patterns' structural dimension through a set of (re-)configuration combinators and constraints that all instances of a pattern must obey. Both types and instances of architectural patterns are semantically represented as bigraphical reactive systems and operations upon them as reaction rules. Such a bigraphical semantics provides a rigorous model for Archery patterns and reduces constraint verification in architectures to a type-checking problem.

1 Introduction

In a number of contexts the term architectural pattern is used as an architectural abstraction. The expression is taken in the usual sense in classical software architecture – a known solution to a recurring design problem. In [4] it is characterised as a description of element and configuration types, and a set of constraints on how to use them. Available catalogs such as [8] provide a vocabulary for their use at a high abstraction level. However, the lack of formality in their pattern documentation prevents its usage for developing precise architectural specifications on top of them, and in consequence, any tool-supported analysis and verification.

Such is the motivation behind Archery, a language to describe the behaviour of pattern elements, a subset of which was recently presented at [13]. Its semantics is given by translation to mCRL2 [10]. A pattern specification in Archery comprises a set of architectural elements (connectors and components) and their associated behaviours. An architecture describes a particular configuration that instances of a pattern’s elements assume. This configuration has an emergent behaviour and constitutes an instance of the pattern. Then, both patterns and elements define the types of behaviour expected from instances. The language supports hierarchical composition of architectures.

This paper, extends Archery to the so-called structural dimension of architectural patterns. This comprises the usage of typed variables to contain and reference instances, a set of scripting operations to build architectural configurations, and a set of primitives to specify constrains over such configurations.
Constraints restrict the class of valid configurations that architectures, instances of a particular pattern, may adopt. Then, reconfigurations are only enabled if respecting the pattern constraints. For instance, a reconfiguration script that connects two clients in a Client-Server architecture violates the intended use of the pattern and should be prevented.

A second contribution of this paper is a semantics for the structural dimension of Archery on top of Bigraphical Reactive Systems (BRS) [11]. The theory of BRSs was developed to study systems in which locality and linking of computational agents varies independently, and to provide a general unifying theory in which existing calculi for concurrency and mobility can be represented. The two main constituents of a BRS are a bigraph and a set of parametric reaction rules. The former specifies the BRS structure as two orthogonal graphs upon the same set of nodes, one modelling locality, and another linking. Rules model its dynamics, i.e., how the structure is reconfigured through reaction.

The theory of BRSs has a precise definition. A bigraph, expressed as a tuple of functions, is an arrow in a category. Its domain and codomain are objects. A more restrictive category can be defined for bigraphs by including in their definition a mechanism, called sorting, that constrains the configurations they can adopt. This setting allows the formal treatment of the encoded system. In particular, if conditions are met [11], it allows to automatically derive a labelled transition system (LTS) from a BRS, in which behavioural equivalence is a congruence.

The choice of BRS as a semantical framework for Archery arose naturally as the language was expected to allow for independently modifying both placing and linking of pattern instances. At a more fundamental level, the structural dimension of patterns and architectures become encoded as arrows in a suitable category\(^3\). Finally, the use of bigraphs reduces the problem of verifying whether an architectural constraint holds for a pattern to a certain kind of type-checking. Actually, once a structural constraint is encoded as a sorting, to check if it is verified by an architecture amounts to translating the latter to a bigraph and prove that such a bigraph belongs to the category defined by the sorting.

The bigraphical encoding presented here is also the basis, along the work in [5], to explore in [12] the automatic derivation of LTS whose states stand for the different configurations the corresponding architecture can adopt. This makes possible to resort to behavioural equivalence to compare the application of different patterns in reconfiguring systems.

The following sections illustrate how Archery can be endowed with a bigraphical semantics. For such purposes we limit ourselves to a subset of the scripting operations and an example constraint. The full version of the language can be found in [12]. The rest of the paper is organised as follows: section 2 introduces the Archery with both extensions. Section 3 briefly recalls the basic theory of BRS and section 4 develops the formal semantics for the structural dimension of Archery. Finally, section 5 concludes and discusses future work.

---

\(^3\) In fact, the name Archery comes from a comment in Steve Awodey’s book [3] emphasising the importance of arrows in category theory: “...the subject might better have been called abstract function theory, or perhaps even better: archery.”
2 The Archery Language

We structure Archery as a core and two extensions, respectively named Archery-Core, Archery-Script, and Archery-Structural-Constraint. The first is a slightly modified version of the language presented in [13], the second adds the operations for building configurations, and the third incorporates the primitives for defining structural constraints. The structure follows the differences in how their semantics are defined. While both behavioural and structural semantics are defined for Archery-Core, only structural semantics are given to Archery-Script and Archery-Structural-Constraint. The three language subsets are endowed structural semantics by translations to bigraphs. However, the codomain of each translation differs, and the third subset requires a more involved approach.

2.1 Archery-Core

A specification in Archery-Core comprises one or more patterns and a main architecture. The first rule of the grammar, shown in Figure 1, indicates this by equating the Spec non-terminal to one or more Pat and a Var non-terminals. Note that several non-terminal are undefined; the grammar leaves out the definition of the ones that are not relevant to the structural dimension.

\[
\begin{align*}
\text{Spec} & ::= \text{Pat}+ \text{Var} \\
\text{Pat} & ::= \text{pattern} \text{TYPEID} (\text{PatPars}? ) \text{elements} \text{Elem}+ \text{end} \\
\text{Elem} & ::= \text{element} \text{TYPEID} (\text{ElemPars}? ) \text{Behaviour} \text{ElemInterface} \\
\text{ElemInterface} & ::= \text{interface} \text{Port}+ \\
\text{Port} & ::= (\text{in}|\text{out}) \text{ID} ; \\
\text{Var} & ::= \text{ID} : \text{TYPEID} = \text{Inst} ; \\
\text{Inst} & ::= (\text{ElemInst}|\text{PatInst}) \\
\text{ElemInst} & ::= \text{TYPEID} (\text{ElemInstPars}? ) \\
\text{PatInst} & ::= \text{architecture} \text{TYPEID} (\text{PatInstPars}? ) \text{ArchBody} \text{end} \\
\text{ArchBody} & ::= \text{Instances} \text{Attachments}? \text{ArchInterface}? \\
\text{Instances} & ::= \text{instances} \text{Var}+ \\
\text{Attachments} & ::= \text{attachments} \text{Att}+ \\
\text{Att} & ::= \text{from} \text{PortRef} \text{to} \text{PortRef} ; \\
\text{ArchInterface} & ::= \text{interface} \text{Ren}+ \\
\text{Ren} & ::= \text{PortRef} \text{as} \text{ID} ; \\
\text{PortRef} & ::= \text{ID,ID}
\end{align*}
\]

Fig. 1: Grammar Fragment for Archery-Core

A pattern is specified according to the rule expanding the Pat non-terminal. Its definition contains, a TYPEID token that represents the identifier for it, an optional list of formal parameters, and one or more architectural elements Elem,
i.e., specified according to the Elem non-terminal. For instance, the specification in Listing 1 includes two patterns: ClientServer and PipeFilter.

Each architectural element in a pattern is specified as described by Elem. Its definition comprises: a TYPEID token as its identifier, an optional list of formal parameters, a description Behaviour of its behaviour, and a description ElemInterface of its interface. The behaviour is specified with a slightly modified subset of mCRL2 limiting its expressivity to sequential processes. Its description must contain one ore more process expressions, as the one shown in line 5, and a list of action definitions, like in line 4. The first process is the initial behaviour of the instance and may call other processes defined within the element. The interface contains one or more ports Port. A port is defined by a direction indicator, either in or out, and an ID token that must match an action name in the list of action definitions. For instance, the interface Server defines two ports in line 6. We adopt the underlying metaphor of water flow in [2] for ports: an in port receives input from any port connected to it, and an out port sends output to all ports connected to it. Ports are synchronous: actually a suitable process algebra expression can be used to emulate any other port behaviour.

Listing 1: Example Patterns and Architectures

```plaintext
pattern ClientServer()
  elements
    element Server()
      act rreq, sres, cres;
      proc Server() = rreq.cres.sres.Server();
    interface in rreq; out sres;
    element Client()
      act prcs, sreq, rres;
      proc Client() = prcs.sreq.rres.Client();
    interface in rres; out sreq;
  end
pattern PipeFilter()
  elements
    element Pipe()
      act accept, forward;
      proc Pipe() = accept.forward.Pipe();
    interface in accept; out forward;
    element Filter()
      act rec, trans, send;
      proc Filter() = rec.trans.send.Filter();
    interface in rec; out send;
  end
cs : ClientServer = architecture ClientServer()
instances
  s1 : Server = architecture PipeFilter()
instances
  f1 : Filter = Filter(); f2 : Filter = Filter();
  p1 : Pipe = Pipe();
```

301
30 from f1.send to p1.accept;
31 from p1.forward to f2.rec;
32 interface f1.rec as rreq; f2.send as sres;
33 end
34 c1 : Client = Client(); c2 : Client = Client();
35 attachments
36 from c1.sreq to s1.rreq; from c2.sreq to s1.rreq;
37 from s1.sres to c1.rres; from s1.sres to c2.rres;
38 end

A variable and its value is defined according to \texttt{Var}. The variable has an ID token as its identifier, followed by a TYPEID token that must match an element or pattern name. The value can be either a pattern \texttt{PatInst} or an element \texttt{ElemInst} instance. Note that the variable that follows the pattern definitions, as indicated in the first grammar rule, and as shown in line 23 of the example, must contain an architecture (the main one).

An architecture defines a set of variables and describes the configuration adopted by the instances in them. It contains: a TYPEID token that must match a pattern name, an optional list of actual arguments, a set of variables \texttt{Var}, an optional set of attachments \texttt{Att}, and an optional interface \texttt{ArchInterface}. Each variable in the set must have as type an element defined in the pattern the architecture is instance of. If the variable has as assigned value an element instance \texttt{ElemInst}, it is defined by a TYPEID and a list of actual parameters. If it has a pattern instance, like between lines 25 and 33 of the example, a nested architecture is defined. Each attachment \texttt{Att} includes a port reference \texttt{PortRef} to an out port, and another to an in port. A port reference is an ordered pair of ID tokens, with the first matching a variable identifier, and the second a port of the variable’s instance. Then, an attachment indicates that the out port communicates with the in port, such as in the case of \texttt{f1.send} with \texttt{p1.accept} in line 30. The architecture interface is a set of one or more port renames \texttt{Ren}. Each port rename contains a port reference and an ID token with the external name for the port. Ports not included in the set are not visible from the outside. Including the same port in an attachment and in the interface is incorrect. An example interface with two renames is shown in line 32.

2.2 Archery-Script

Archery-Script is used to specify a script for creating an architecture or for reconfiguring an existing one. It assumes the existence of a process that triggers a scripts under some conditions. Its operations (informally described in Table 1), are defined independently of any pattern. The design principles of patterns are enforced through constraints, as it is shown in Section 2.3. This independence, and the fact that a variable may contain an instance whose type may not necessarily match the variable’s type, allows the reuse of a script in an open family of patterns (related by some refinement relation). We illustrate the operations through the example in Listing 2.
Table 1: Set of Operations in Archery-Script

<table>
<thead>
<tr>
<th>Name</th>
<th>Format</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Import</td>
<td>import(s)</td>
<td>Receives as a parameter a reference s to an Archery specification and imports it to the environment of the executing script (e.g., line 2 in Listing 2).</td>
</tr>
<tr>
<td>Create Variable</td>
<td>v:type</td>
<td>Creates a variable with name v and type type (line 3).</td>
</tr>
<tr>
<td>Create Instance</td>
<td>v=type()</td>
<td>Creates a new instance of type type and assigns it to a variable v (line 4).</td>
</tr>
<tr>
<td>Add Instance</td>
<td>addInst(a,v)</td>
<td>Adds a variable v and the instance in it, to the architecture in variable a (line 5).</td>
</tr>
<tr>
<td>Attach</td>
<td>attach(vf.pf, vt.pt)</td>
<td>Attaches the port pf of the instance in variable vf to the port pt of the instance in variable vt (line 8).</td>
</tr>
<tr>
<td>Deattach</td>
<td>deattach(vf.pf, vt.pt)</td>
<td>Removes the attachment between the port pf of the instance in variable vf and the port pt of the instance in variable vt (line 6).</td>
</tr>
<tr>
<td>Move</td>
<td>move(vs, vt)</td>
<td>Moves the instance in variable vs to the variable vt (line 11). The reference to the contents of vt are lost, but its attachments remain.</td>
</tr>
</tbody>
</table>

The example is divided in three parts and assumes the existence of an initial configuration we call $c_{\text{initial}}$. The configuration is similar to the one in Listing 1, but differs in that the nested architecture (between lines 25 and 33) is replaced by a Server instance (in a single line $s_1: \text{Server}=\text{Server}();$).

The first part of the example reconfigures $c_{\text{initial}}$ by adding and connecting a second server. It starts with an import operation that leaves the configuration in variable cs. The operations in lines 3 and 4, create a new variable s2 and assign a fresh instance of Server to it. Upon that, s2 is included in the architecture in cs. Then the operations in the next two lines remove the attachments among the instances in variables cs.c2 and cs.s1. Subsequently, new attachments are created between the instance in variable cs.c2 with the instance in variable cs.s2. We will refer to the obtained configuration as $c_{\text{first}}$.

Listing 2: Example Script

```plaintext
script
import("initial"); // first part
s2 : Server;
s2 = Server();
addInst(cs, s2);
deattach(cs.c2.sreq, cs.s1.rreq);
deattach(cs.c2.rres, cs.s1.sres);
attach(cs.c2.sreq, cs.s2.rreq);
attach(cs.c2.rres, cs.s2.sres);
import("pf"); // second part
move(pf, cs.s2);
c3 : Client = Client(); // third part
```
addInst(cs, c3);

detach(cs.c2.sreq, cs.s2.rreq);

detach(cs.c2.rres, cs.s2.sres);

attach(cs.c2.sreq, cs.c3.rres);

attach(cs.c2.rres, cs.c3.sreq);

end

The second part of the example starts in the line 10 and shows how a server is replaced. It assumes the existence of a configuration $pf$, similar to the one described between the lines 25 and 33 in Listing 1, but with the architecture contained in a variable named $pf$ of type $PipeFilter$. The script imports the configuration $pf$, and then the instance in $pf$ is moved to the variable $cs.s2$. The instance in the variable $cs.s2$ is now the architecture of type $PipeFilter$ but connected as it was the previous instance in the variable.

The third part begins upon line 12. It creates a new client and connects it in a wrong way. A new variable $c3$ is created and a new instance of the type $Client$ is assigned to it line 12. Next, the fresh variable is included in the architecture in $cs$. Subsequently, the attachments between the instances in variables $cs.c2$ and $cs.s2$ are removed. Then, the script creates two attachments between instances in variables $cs.c3$ and $cs.c2$. The resulting configuration violates the design principle behind a client-server architecture by connecting two clients. We refer to the configuration obtained upon the script execution as $cs_{wrong}$.

2.3 Archery-Structural-Constraint

To rule out configurations such as $cs_{wrong}$, entails the need for mechanisms to constrain what may count as valid instances of a pattern. Since the variable $cs$ in the script of Listing 2 is of type $ClientServer$, we could add to the pattern specification a constraint $\varphi$ to express that clients can only connect to servers and vice versa. We define $\varphi$ for all attachments $att$ in an architecture of type $ClientServer$ as follows

$$client(\text{from}(att)) \Rightarrow server(\text{to}(att)) \land client(\text{from}(att)) \Rightarrow server(\text{to}(att))$$

with $\text{from}$ (respectively, $\text{to}$) a function that returns the variable with the $\text{out}$ (respectively, $\text{in}$) port in $att$, and with $\text{client}$ (respectively, $\text{server}$) a predicate yielding true when its argument is of type $Client$ (respectively, $Server$).

By constraining patterns in this way, we can prevent an operation in a script that generates an invalid configuration. Clearly, $cs_{wrong}$ does not satisfy it. In contrast, the configuration $cs_{first}$ does. Given a configuration $c$ and a constraint $\varphi$, the satisfaction problem can be formulated as $c \models \varphi$, which can be rendered as a type checking assertion in the bigraphical semantics for Archery. Such is the topic of the following sections.

3 Bigraphical Reactive Systems

A Bigraphical Reactive System (BRS) is an inhabitant of a category. The operations and the elementary bigraphs in such category enable an algebraic treatment
of BRs. In the next sections we briefly describe the notions of bigraphs, their algebra, and the parametric reaction rules that make them dynamic. We refer the reader to [11] for more detail on these notions and their precise definitions.

### 3.1 Bigraphs

A bigraph contains a set of nodes related through a parent-child relationship and through edges. The former gives rise to a forest structure called place graph, in which the roots of the trees are the nodes without parent. The latter defines a hypergraph called link graph: a node is related to others by an edge, if each one has a port linked to an end of such edge. A bigraph is said to be concrete if its nodes and edges have identity, and abstract if they not. Figure 2 shows the structure of bigraphs following the anatomy style used in [11]. The abstract bigraph in it has a forest with two trees and a hypergraph with two edges.

![Fig. 2: Anatomy of Bigraphs](image)

The encoding of a system is enabled by the basic signature of a bigraph. Every node has an associated control from a set \( K \) that distinguishes its kind of contribution to the encoding. The control also establishes the number of ports the node has with an arity function \( ar: K \rightarrow N \). The tuple \( (K, ar) \) is the basic signature of a bigraph and in the case of our example \( K = \{L: 2, M: 3\} \).

New bigraphs can be built from existing ones by plugging one into another. The interface of a bigraph defines the form of the bigraphs it can contain – inner face, and the form that a container must accept – outer face. Suppose we divide a bigraph into two parts. A division in a tree leaves a site in one part, and a new root on the other. A division in an edge generates two open links: one called inner name and another called outer name. The roots and outer names are the
outer face, and the sites and inner names the inner face of a bigraph. Figure 2 shows the graphic conventions to depict them.

The category in which a bigraph lives depends on whether it is abstract or not and the signature $K$ over which it is defined. An abstract bigraph becomes an arrow $F : I \to J$ in a category $\text{Bg}(K)$. Its domain $I$ and codomain $J$ are objects in such category. The domain is a tuple $I = \langle n, X \rangle$, in which $n$ is a set of ordinals $\{0, 1, \ldots, n - 1\}$ that index its sites, and $X$ is its set of inner names. Similarly, the codomain is a tuple $J = \langle m, Y \rangle$ with $m$ indexing its roots, and $Y$ its set of outer names. If the bigraph is concrete, the space is a precategory $\text{Bc}(K)$ instead. The reason for using a precategory is that composition is not always defined when nodes and edges have identity.

Undesired arrangements of controls can be ruled out by defining a sorting $\Sigma = (\Theta, K, \Phi)$. The controls in $K$ are classified in a set of sorts $\Theta = \{\theta_0, \ldots, \theta_n\}$, and valid arrangements of sorts are restricted with a formulation rule $\Phi$. The sorts can be assigned to the controls – place sorting, or to the links according to the ports in controls – link sorting. Abstract (respectively, concrete) bigraphs over a sorting $\Sigma$ inhabit a category $\text{Bc}(\Sigma)$ (respectively, precategory $\text{Bc}(\Sigma)$).

### 3.2 Algebra

All bigraphs can be built from elementary ones by applying three basic operations: composition, product and identities. The composition $G \circ F : I \to K$, also denoted $G \cdot F$, of two bigraphs $F : I \to J$ and $G : J \to K$, represents a new bigraph obtained by plugging $F$ into $G$. This operation is only defined when the inner face of $G$ matches the outer face of $F$. The set $|F|$ of node and edge identifiers of $F$ needs to be disjoint with $|G|$ if bigraphs are concrete. When $G \circ F$ is defined, we say that $G$ is a context for $F$. The product of two bigraphs $F_i : \langle m_i, X_i \rangle \to \langle n_i, Y_i \rangle$ ($i = 0, 1$), is a new bigraph $F_0 \mathbin{\otimes} F_1 : \langle m_0 + m_1, X_0 \uplus X_1 \rangle \to \langle n_0 + n_1, Y_0 \uplus Y_1 \rangle$, (with $\uplus$ the union of disjoint sets) that represents placing $F_0$ besides $F_1$. $|F_0| \cap |F_1| = \emptyset$ also needs to hold for concrete bigraphs. The identity bigraph (arrow) of an interface (object) $I = \langle m, X \rangle$ is a tuple $\langle id_m, id_X \rangle$. In practice, a set of derived operations defined on top of the basic ones and elementary bigraphs is actually used.

The elementary bigraphs that do not have nodes are divided in the ones that only have roots and sites – placings ($\phi$), and the ones that only have (outer and inner) names – linkings ($\lambda$). Placings can be generated from three elementary forms: a root with no sites $1 : 0 \to 1$; a symmetry $\gamma_{1,1} : 2 \to 2$ that exchanges the indexes of roots with the ones of sites; and a join $\text{join} : 2 \to 1$ of two sites into one root. A merge bigraph can be derived as $\text{merge}_{n+1} = \text{join} \circ (id_1 \otimes \text{merge}_n)$. Similarly, linkings can be generated from two elementary forms: the substitution $\gamma_X/Y$ of a set of names $X$ with one name $y$; and the closure $/x$ of a link $x$. The only elementary bigraph that introduces nodes is $K_\mathcal{X} : n \to (1, \{x\})$, defined for each control $K : n$ (with $n$ ports), gives rise to a bigraph with a single node whose $n$ ports are bijectively linked to $n$ names in $\mathcal{X}$.

Some abbreviations for operations we may use are as follows: we may write $F \circ G$ instead of $(F \otimes id_I) \circ G$ when there is no ambiguity; given a linking
\( \lambda : Y \rightarrow Z \) and a bigraph \( G : I \rightarrow \langle m, X \rangle \) with \( Y = X \cup X' \), we may write \( \lambda \circ G \) instead of \((id_m \otimes \lambda) \circ (G \otimes X')\) when \( m \) and \( X \) are clear from the context.

The derived operations are: parallel product, nesting and merge product. The parallel product of two bigraphs \( F_i : \langle m_i, X_i \rangle \rightarrow \langle n_i, Y_i \rangle \) \((i = 0, 1)\) is defined as \( F_0 \parallel F_1 : \langle m_0 + m_1, X_0 \cup X_1 \rangle \rightarrow \langle n_0 + n_1, Y_0 \cup Y_1 \rangle \), a tensor product of the two bigraphs, with the exception that the link map allows name sharing. The result of the nesting of two bigraphs \( F : I \rightarrow \langle m, X \rangle \) and \( G : m \rightarrow \langle n, Y \rangle \) that may share names is a bigraph \( G.F : I \rightarrow \langle n, X \cup Y \rangle \) defined by the expression \((id_n \parallel G) \circ F\).

The merge product of two bigraphs \( G_i \) \((i = 0, 1)\) is \( \text{merge} \circ \left( G_0 \parallel G_1 \right) \), i.e., the merge of the parallel product of them.Abbreviations that we may use are as follows: \( y/x \circ G \) instead of \((y/x \parallel id_I) \circ G\) with \( I = \langle n, Z \rangle \), when \( G \) has outer face \( \langle n, X \sqcup Z \rangle \); \( A \) for the bigraph \( A.1 \) when the control \( A \) has no children.

The algebraic expression in Figure 2 represents the bigraph shown above it, and is defined in terms of these elementary bigraphs and operations.

### 3.3 Reaction Rules

A parametric reaction rule is a tuple \( \langle R : m \rightarrow J, R' : m' \rightarrow J', \eta \rangle \), with \( R \) and \( R' \) bigraphs respectively called redex and reactum, and \( \eta \) an instantiation map. \( R \) and \( R' \) cannot have edges that are not connected to any port or inner name, \( R' \) cannot have barren roots nor have names that are not linked. The instantiation map assigns to each ordinal in \( m' = \{0, 1, \ldots, m' - 1\} \) an ordinal \( m = \{0, 1, \ldots, j, \ldots, m - 1\} \). When a bigraph \( F \) matches the redex, it is replaced with the reactum. The sites in \( F \) are placed in the sites of the reactum according to \( \eta \). If we name the bigraphs contained by \( F \) according to the sites \( m \) in the reactum in which they are placed, we obtain a sequence \( d_0, d_1, \ldots, d_j, \ldots, d_m \). Then, the expression \( \eta(i) = j \) tells that \( d_j \) will be placed in the \( i^\text{th} \) site of the reactum.

Bigraphs that have an associated set of reaction rules are defined over a dynamic signature. It differs from the basic in that each control is assigned one of the three values as follows: atomic – for controls of nodes without children (barren), active – for non-atomic controls that allow reactions to occur among the nodes inside, passive – for non-atomic and non-active controls. A reaction only takes place if the bigraph matching the redex is in an active context, i.e., in a root, or in an active node with all ancestors active as well.

The abstract (respectively concrete) BRS with sorting \( \Sigma \) and parametric reaction rules \( \mathcal{R} \) live in a category \( \text{Bg}(\Sigma, \mathcal{R}) \) (\( \text{Bg}(\Sigma, \mathcal{R}) \)).

### 4 Bigraphical Modelling of Archery Specifications

In this section we provide a bigraphical semantics for Archery. We respectively translate Archery-Core and Archery-Script specifications into bigraphs in categories \( \text{Bg}(\Sigma_{ Arch-Core}, \mathcal{R}_{ Arch-Core}) \) and \( \text{Bg}(\Sigma_{ Arch-Script}, \mathcal{R}_{ Arch-Script}) \). Since each Archery-Structural-Constraint constraint generates a different category, we limit to define \( \text{Bg}(\Sigma_\varphi, \mathcal{R}_{ Arch-Core}) \) for the example constraint \( \varphi \) described in Section 2.3 and leave a generic method to [12].
4.1 Archery-Core

Function $T$ (1) translates an Archery-Core specification into a bigraph in category $\mathcal{B}g(\Sigma_{Arch-Core}, R_{Arch-Core})$. It takes a $Spec$ and returns the parallel product of bigraphs that result of translating each $Pat$ in $Pat^+$, and a variable $Var$ containing the main architecture. We describe the signature and rules as we explain the translation, and leave the sorting for the end of the section. Table 2 lists the controls in $\Sigma_{Arch-Core}$ and the sort assignment to their ports, and Table 3 the rules in $R_{Arch-Core}$. Figure 3 shows the bigraph that returns the application of function $T$ to the pattern ClientServer in Listing 1, and Figure 4 to the architecture between lines 25 and 33.

\[
T(Spec) = \bigparallel_{Pat^+} T(Pat) \parallel T(Var) \tag{1}
\]

\[
T(Pat) = Pat_{TYPEID}.\bigparallel_{Elem^+} T(Elem) \tag{2}
\]

\[
T(Elem) = Elem_{TYPEID}.\bigparallel_{Port^+} T(Port) \tag{3}
\]

\[
T(in ID) = NewIn_{ID}, T(out ID) = NewOut_{ID} \tag{4}
\]

\[
T(Var) = T(Var, 1) \tag{5}
\]

\[
T(Var, B) = NewVar_{TYPEID}(T(Inst, ID, B)) \tag{6}
\]

\[
T(PatInst, idVar, B) = NewInst_{TYPEID, idVar}.(\bigparallel T(idVar, Var^+, Att^*, Ren^*, B)) \tag{7}
\]

\[
T(idVar, Var, Var^+, Att^*, Ren^*, B) = T(idVar, Var^+, Att^*, Ren^*, B)) \tag{8}
\]

\[
T(idVar, Att^*, Ren^*, B) = T(Att^*, Ren^*, B) \tag{9}
\]

The translation of a pattern $Pat$ (2) creates a node $Pat$ (a node with control $Pat$) and outer name $TYPEID$, and nests the merge product of translating each of its elements. It translates each element $Elem$ (3) into a node $Elem$ with outer name $TYPEID$, and nests the merge product of translating each of its ports. The translation of each port $Prt$ yields a node $NewIn$ if its direction is in, and a node $NewOut$ if it is out. In both cases, the node has $ID$, the identifier of the port, as outer name.

Function $T$ translates a variable $Var$ (5) into a node $NewVar$ with outer names $ID$ and $TYPEID$. Then, it nests its recursive call with $Inst, ID,$ and
Table 2: Sorting for Archery-Core

<table>
<thead>
<tr>
<th></th>
<th>Ctrl</th>
<th>Arity</th>
<th>Activeness</th>
<th>Sorts</th>
<th>Represented Item</th>
</tr>
</thead>
<tbody>
<tr>
<td>Pat</td>
<td>1</td>
<td>passive</td>
<td>u</td>
<td>A pattern</td>
<td></td>
</tr>
<tr>
<td>Elem</td>
<td>1</td>
<td>passive</td>
<td>u</td>
<td>An element</td>
<td></td>
</tr>
<tr>
<td>Newln</td>
<td>1</td>
<td>passive</td>
<td>u</td>
<td>An in port within an element definition</td>
<td></td>
</tr>
<tr>
<td>In</td>
<td>1</td>
<td>atomic</td>
<td>i</td>
<td>An in port within an instance</td>
<td></td>
</tr>
<tr>
<td>NewOut</td>
<td>1</td>
<td>passive</td>
<td>u</td>
<td>An out port within an element definition</td>
<td></td>
</tr>
<tr>
<td>Out</td>
<td>1</td>
<td>atomic</td>
<td>o</td>
<td>An out port within an instance</td>
<td></td>
</tr>
<tr>
<td>NewInst</td>
<td>2</td>
<td>passive</td>
<td>uu</td>
<td>Instance creation and assignment</td>
<td></td>
</tr>
<tr>
<td>Inst</td>
<td>1</td>
<td>active</td>
<td>u</td>
<td>An Instance</td>
<td></td>
</tr>
<tr>
<td>NewVar</td>
<td>2</td>
<td>passive</td>
<td>uu</td>
<td>Variable creation</td>
<td></td>
</tr>
<tr>
<td>Var</td>
<td>2</td>
<td>active</td>
<td>uu</td>
<td>A variable</td>
<td></td>
</tr>
<tr>
<td>AddVar</td>
<td>2</td>
<td>passive</td>
<td>uu</td>
<td>Movement of one variable into another</td>
<td></td>
</tr>
<tr>
<td>NewAtt</td>
<td>5</td>
<td>passive</td>
<td>uuuu</td>
<td>Attachment creation</td>
<td></td>
</tr>
<tr>
<td>NewRen</td>
<td>3</td>
<td>passive</td>
<td>uuu</td>
<td>Rename creation</td>
<td></td>
</tr>
<tr>
<td>From</td>
<td>2</td>
<td>atomic</td>
<td>fu</td>
<td>Attachment end for out port</td>
<td></td>
</tr>
<tr>
<td>To</td>
<td>2</td>
<td>atomic</td>
<td>tu</td>
<td>Attachment end for in port</td>
<td></td>
</tr>
</tbody>
</table>

Fig. 3: Bigraph for the Client-Server Pattern

a bigraph $B$ as parameters. NewVar$_{ID,TYEID}$ partially matches the redex of Rule 1 (in Table 3). The reaction yields a bigraph with two roots: one with a node Var$_{ID,TYEID}$ that represents the variable, and another with the former contents, matching parameter $d_0$, of the node NewVar.

The translation of an instance (6) also takes as parameters, the id idVar of the variable that contains it, and a bigraph $B$. The instance can either be an element or a pattern instance, with the form ElemInst or PatInst, respectively. In both cases a node NewInst is created with the type TYPEID of the instance and idVar as outer names. In the former case, $B$ is nested. In the latter, the result of translating the three parts of the architecture: its instances in variables Var+, its attachments Att+, and its interface Ren+, is nested.

NewInst$_{TYEID,idVar}$ triggers Rule 2 if TYPEID refers to an element, and Rule 3 if it refers to a pattern. The two rules create a node Inst$_{TYEID}$, and place it inside the node Var$_{idVar}$ (already created). NewInst$_{TYEID,idVar}$ disappears.
Table 3: Parametric Reaction Rules for Archery-Core

<table>
<thead>
<tr>
<th>Rule</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>New Variable: NewVar_x, d_0 \rightarrow Var_x, I \parallel d_0</td>
</tr>
<tr>
<td>2</td>
<td>Create Element: Elem_x, d_0 \parallel Var_x, I \parallel NewInst_x, d_1 \rightarrow</td>
</tr>
<tr>
<td>3</td>
<td>Create Pattern: Pat_x, d_0 \parallel Var_x, I \parallel NewInst_x, d_1 \rightarrow</td>
</tr>
<tr>
<td>4</td>
<td>Create In Port: Inst_x (NewIn_x</td>
</tr>
<tr>
<td>5</td>
<td>Create Out Port: Inst_x (NewOut_x</td>
</tr>
<tr>
<td>6</td>
<td>Add Instance: Var_x, I (Inst_x</td>
</tr>
<tr>
<td>7</td>
<td>Add Attachment: Var_x, I (Inst_x (Out_x</td>
</tr>
<tr>
<td>8</td>
<td>Add Rename Out: Var_x, I (Out_x</td>
</tr>
<tr>
<td>9</td>
<td>Add Rename In: Var_x, I (ln_x</td>
</tr>
</tbody>
</table>

and its contents are combined with Var\_id\_var using parallel product. Rule 3 also copies the contents of Elem\_YPEID into Inst\_YPEID. Since the context for the contents is active (a Inst with active ancestors) Rules 4 and/or 5 are triggered. Rule 4 transforms a node NewIn into a node In, keeping its outer name, but also closing it. Rule 5 does a similar work but for nodes NewOut and Out. These three rules are needed to represent the creation of an element instance. A single rule cannot achieve the desired effect because of how the copy of a closure operation is treated (see chapter eight of [11]) by parametric reaction rules.

Function \( T \) (T) recursively translates the list of variables Var in the architecture. It translates the first variable with a recursive call that takes two parameters: the variable Var; and the translation of the rest of the architecture nested in a node AddVar\_id\_Var, ID, with id\_Var the id of the variable that contains the architecture, and ID the identifier of Var. When the list is empty, the translation continues with the attachments Att\* and the renames Ren\*.

The effect of Rule 6, that is partially triggered by AddVar\_id\_Var, ID, is to make the variable with identifier ID, a variable of the architecture in variable id\_Var. It produces a bigraph with two roots: one contains Var\_ID, nested inside the node with control Inst that represents the architecture; and the other holds the former contents of AddVar\_id\_Var, ID.

In a similar way as with variables, Function \( T \) (8) processes the list of attachments Att. It takes the header, creates a node with control NewAtt, and nests the bigraph obtained with its recursive call with the tail. An Att specification has four identifiers: id\_IF and id\_IT that respectively reference the instances with the out and in ports, and id\_PF and id\_PT that correspond to the out and in ports. The node is created using these four identifiers and a value att of the call to the function uniqueId as outer names. Function uniqueId provides a unique
identifier for the node representing an attachment creation. When the function receives an empty list, it begins with the translation of renames \( \text{Ren} \).

\( \text{NewAtt}_{id_{IF}, id_{PF}, id_{IT}, id_{PT}, att} \) partially matches Rule 7. Its first and third outer names respectively link it with the variables containing the origin \( Var_{id_{IF},-} \) and the end \( Var_{id_{IT},-} \) instances of the communication, and its second and fourth match their participating ports, \( \text{Out}_{id_{PF}} \) and \( \text{In}_{id_{PT}} \). The rule removes the node, adds two new, and creates three edges. It adds \( \text{From}_{id_{PF},att} \) inside the instance of \( Var_{id_{IF},-} \), and \( \text{To}_{id_{PT},att} \) inside the instance of \( Var_{id_{IT},-} \). This creates three edges: one closed edge between \( \text{Out}_{id_{PF}} \) and \( \text{From}_{id_{PF},att} \), another between \( \text{In}_{id_{TF}} \) and \( \text{To}_{id_{TF},att} \), and an open edge between \( \text{From}_{id_{PF},att} \) and \( \text{To}_{id_{TF},att} \). It defines a route – a sequence of nodes and edges, between the ports.

Function \( T \) (9) recursively processes a list of renames \( \text{Ren}^* \). Each \( \text{Ren} \) has an identifier \( id_{Inst} \) for the instance that contains the port, an identifier \( id_{Prt} \) for such port, and a new name \( id_{New} \) for it. The function creates a \( \text{NewRen}_{id_{Inst}, id_{Prt}, id_{New}} \) for each \( \text{Ren} \).

\( \text{NewRen}_{id_{Inst}, id_{Prt}, id_{New}} \) partially matches either Rule 8 or 9, depending on whether the control of the node with outer name \( id_{Prt} \) respectively is \( \text{Out} \) or \( \text{In} \). The two rules perform a similar task, they remove the node triggering the Rule, place its contents in a parallel root, and add a substitution \( id_{New}/\{id_{Prt}\} \).

The sorts for the links of a control \( \Theta = \{o, f, t, i, u\} \) and the formulation rule \( \Phi \) ensure valid configurations representing attachments: routes can only connect ports with opposite direction (\( \text{in/out} \)). Formulation rule \( \Phi \) restricts the structure as follows: a link with a point \( o \) (port or inner name with sort \( o \)) can only have other points \( f \); a link with a point \( i \) can only have other points \( t \); a link with a point \( u \) has sort \( u \) and no constraints. The sorting assignment in Table 2 and \( \Phi \) prevents a bigraph representing attachments between two ports with the same direction. Figure 4 shows two edges between \( \text{Out} \) (respectively, \( \text{In} \)) and \( \text{From} \) (\( \text{To} \))
nodes that satisfy $\Phi$. It does not show sorts $u$ with the exception of nodes From and To. An attachment, for instance, between two $In$ nodes violates $\Phi$ since it would require an edge with a point $i$ and a point $f$.

4.2 Archery-Script

We translate a script into a bigraph in $BG(\Sigma_{Arch-Script}, \mathcal{R}_{Arch-Script})$. Both the sorting and the parametric reaction rules extend the ones defined for Archery-Core. $\Sigma_{Arch-Script}$ includes two more controls and $\mathcal{R}_{Arch-Script}$ includes the two parametric reaction rules in Table 4.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
<th>Rule</th>
</tr>
</thead>
<tbody>
<tr>
<td>10 Remove Attachment</td>
<td>$Var_{d,\ldots}({Inst_{\ldots}(Out_{pt} \mid d_0)} \mid From_{pt,att} \mid d_1)$</td>
<td>$\rightarrow$</td>
</tr>
<tr>
<td>11 Move Instance</td>
<td>$Var_{vd,\ldots}(Inst_{\ldots}(d_1)) \parallel Var_{vd,\ldots}(d_2) \parallel Move_{inst_{vd,\ldots}d_3}$</td>
<td>$\rightarrow$</td>
</tr>
</tbody>
</table>

Function $T_S$ carries out the translation of a script $t = [t_1 \ t_2 \ \ldots \ t_n]$ by processing the first operation and returning a combination of the result and the recursive call with the tail of the sequence. Each operation $t_i$ has as type one of the seven in Table 1. Expression (10) translates an import operation into the parallel product of the application of $T$ to the specification $Spec$, and the recursive call with the rest of the script. Expressions (11) to (16) translate $t$ by nesting the translation of the tail of $t$ in a node that results from translating $t_1$. The created node partially triggers one of the reaction rules in $\mathcal{R}_{Arch-Script}$.

We introduce the controls and rules in expressions (15) and (16) since they are not present in $\Sigma_{Arch-Core}$ and $\mathcal{R}_{Arch-Core}$. The former creates a node with control RemAtt that represents a remove attachment operation. It is a passive control with four ports of sort $u$. The outer names of the node are the source variable and port names (respectively, $vf$ and $pf$), and the corresponding target names (respectively, $vt$ and $pt$). This node partially triggers Rule 10, that removes the nodes RemAtt, From, and To, making the edge representing the attachment disappear. It also places the contents of the first, that match the parameter $d_4$ in a parallel root. The latter (16) creates a node MoveInst that represents instance movement operations. The control is passive and has two ports with sort $u$: one identifier $vo$ representing the original container for the instance, and another $vd$ for the container to where it is moved. The node partially matches the redex of Rule 11. The reaction nests the contents of $Var_{vo,\ldots}$, matching $Inst_{\ldots}(d_1)$, into $Var_{vd,\ldots}$. The former contents of the destination are lost. The original variable keeps the contents matching $d_2$ (outside the instance), and the contents matching $d_3$ are place in a parallel root.
\[
\begin{align*}
T_S([\text{import}(\text{Spec}); \, t]) &= T(\text{Spec}) \parallel T_S(t) \\
T_S([v : \text{type}; \, t]) &= \text{NewVar}_v.\, \text{type}.\, T_S(t) \\
T_S([v = \text{type}(); \, t]) &= \text{NewInst}_v.\, \text{type}.\, T_S(t) \\
T_S([\text{addInst}(a, v); \, t]) &= \text{AddVar}_a.\, v.\, T_S(t) \\
T_S([\text{attach}(v_f, pf, vt, pt); \, t]) &= \text{NewAtt}_{v_f, pf, vt, pt, uniqueId}.\, T_S(t) \\
T_S([\text{deattach}(v_f, pf, vt, pt); \, t]) &= \text{RemAtt}_{v_f, pf, vt, pt}.\, T_S(t) \\
T_S([\text{move}(v_o, v_d); \, t]) &= \text{MoveInst}_{v_o, v_d}.\, T_S(t) \\
T_S([\,]) &= 1
\end{align*}
\]

4.3 Archery-Structural-Constraint

The way constraints are dealt within the bigraphical framework discussed in this paper is now illustrated through an example. Let us consider the constraint \( \varphi \) formulated in Section 2.3. We derive from it a place sorting \( \Sigma_\varphi \). Note that, in general, this derivation can be automated [12]. Then, a specification that fulfils \( \varphi \) is translated to a bigraph in \( B_g(\Sigma_\varphi, \mathcal{R}_{Arch-\text{Core}}) \).

For this example, we define the set of sorts as \( \Theta = \{ \text{cli, ser, att, oth} \} \). The sort of a node \( \text{Var}_{-\text{type}} \) depends on its outer name \( \text{type} \): if \( \text{type} \) is \( \text{Client} \) it has sort \( \text{cli} \), if \( \text{type} \) is \( \text{Server} \), \( \text{ser} \). Nodes \( \text{From} \) and \( \text{To} \) have sort \( \text{att} \), and nodes with other controls sort \( \text{oth} \). The formation rule \( \Phi \) is as follows: a node \( \text{att} \) immediately in a node \( \text{cli} \) can only have an edge to an \( \text{att} \) immediately in a node \( \text{ser} \). Given two nodes \( w \) and \( w' \), \( w \) is in \( w' \) if the former has \( w' \) as ancestor in the parent-child relationship.

It can now be verified whether a specification \( \text{Var} \) representing an instance of type Client-Server preserves constraint \( \varphi \), by checking if the type of bigraph \( T(\text{Var}) \) is \( B_g(\Sigma_\varphi, \mathcal{R}_{Arch-\text{Core}}) \). In Section 2.2 we described \( cs_{\text{first}} \) and \( cs_{\text{wrong}} \) as two configurations. Figure 5 partially shows the bigraphs that encode them. Only the sorts \( \text{att, cli and ser} \), and nodes that participate in attachments are shown. Control labels, outer names and nodes not relevant for checking \( \Phi \) are not included. Figure 5a contains a bigraph that partially encodes \( cs_{\text{first}} \). It can be observed that all four nodes \( \text{att} \) in \( \text{cli} \) (respectively, \( \text{ser} \)) only have edges to nodes \( \text{att} \) in nodes \( \text{ser} \) (respectively, \( \text{cli} \)). Then, the bigraph is \( B_g(\Sigma_\varphi, \mathcal{R}_{Arch-\text{Core}}) \) and configuration \( cs_{\text{first}} \) satisfies \( \varphi \). In contrast, the encoding of \( cs_{\text{wrong}} \) shown in Figure 5b, does not fulfil formation rule \( \Phi \): the nodes \( \text{att} \) in the node \( \text{cli} \) with outer name \( c_1 \), has edges with nodes \( \text{att} \) in another node \( \text{cli} \). Therefore, the bigraph is not \( B_g(\Sigma_\varphi, \mathcal{R}_{Arch-\text{Core}}) \).

5 Conclusions

In this paper we introduced Archery, a modelling language for software architectural patterns rooted in the process algebra trend [10]. The language allows the specification of both structural and behavioural dimensions of architectures.
(Archery-Core), scripts to (re)configure such architectures (Archery-Script), and constraints to ensure that they obey the design principles of the pattern they are instance of (Archery-Structural-Constraint).

A second contribution of the paper was the development of a bigraphical semantics for Archery. To respect space limits, this was fully presented for Archery-Core, partially for the scripting component and illustrated through an example for constraints. By doing so, we were able to reduce the constraint satisfaction verification to a type checking problem.

We can distinguish two approaches in the design of languages that provide support for both the behavioural and structural dimensions, in architectural design. One is to extend a structure-based language with a behavioural model [6], and the other is to build the architectural language on top of the behavioural model [1], by upgrading it with architectural constructs. Our work is along the lines of the latter approach but with the difference that we used bigraphs as a foundation for the structural dimension. Benefits of using the bigraphical theory include its solid categorical framework, its independent treatment of locality and linking of computational agents, and its role of unifying theory for concurrency and mobile calculi. The work in [9] also provides a bigraphical semantics to an architectural description language. While our encoding uses a single signature to encode any pattern, theirs requires different signatures for different patterns. There are two main approaches to the reconfiguration of pattern instances: one is to define a generic set of operations and reflect a pattern’s design principles with constraints that prevent illegal configurations; and another is to design a pattern-specific set of operations that allow to correctly (re)configure instances [7]. Our work is aligned with the former.

Fig. 5: Bigraphs for Example Configurations

(a) $c_{s_{\text{first}}}$
(b) $c_{s_{\text{wrong}}}$
As part of future work we mention the derivation process for sortings that encode constraints. The process must ensure that the resulting sorting does not prevent the automatic derivation of an LTS for a BRS, and consider the decidability and complexity of type-checking.

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References

A formal framework for coordinated simulation of heterogeneous service-oriented applications

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Abstract. Early design and validation of service-oriented applications is hardly feasible due to their distributed, dynamic, and heterogeneous nature. In order to support the engineering of such applications and discover faults early, foundational theories, modeling notations and analysis techniques for component-based development should be revisited. This paper presents a formal framework for coordinated simulation of service-oriented applications based on the OSOA open standard Service Component Architecture (SCA) for heterogeneous service assembly and on the formal method Abstract State Machine (ASM) for modeling notions of service behavior, interactions, and orchestration in an abstract but executable way. The proposed framework is exemplified through a Robotics Task Coordination case study of the EU project BRICS.

1 Introduction

Service-oriented applications are playing so far an important role in several application domains (e.g., information technology, health care, robotics, defense and aerospace, to name a few) since they offer complex and flexible functionalities in widely distributed environments by composing, possibly dynamically “on demand”, different types of services. Web Services is the most notable example of technology for implementing such components. On top of these service-oriented components, business processes and workflows can be (re-)implemented as composition of services – service orchestration or service coordination³. Examples of composition languages are WS-BPEL⁴, XLANG⁵, and Jolie [30], to name a few.

This emerging paradigm raises a bundle of problems, which did not exist in traditional component-based design, where abstraction, encapsulation, and modularity were the main concerns. Early designing, prototyping, and testing of the

³ Thorough this paper, the terms coordination and orchestration are interchangeable.
⁴ www.oasis-open.org
⁵ www.ebpml.org/xlang.htm
functionality of such assembled service-oriented applications is hardly feasible since services are discoverable, loosely-coupled, and heterogeneous (i.e. they differ in their implementation/middleware technology) components that can only interact with others on compatible interfaces. Concurrency and coordination aspects [4] that are already difficult to address in component-based system design (though extensively studied), are even more exacerbated in service-oriented system design. Components encapsulate and hide to the rest of the system how computations are ordered in sequential threads and how and when computations alter the system state. The consequence of improper management of the order and containment relationships or the total absence of an explicit coordination model in a complex, concurrent system leads to deadlock and starvation [18].

In order to support the engineering of service-oriented applications, to discover faults early, and to improve the service quality (such as efficiency and reliability), foundational theories and high-level formal notations and analysis techniques traditionally used for component-based systems should be revisited and integrated with emerging service development technologies. In the Robotics context, in particular, as the Internet is leveraged to connect humans to robots and robots to the physical world, there is a strong requirement to investigate service-oriented engineering approaches and knowledge representations to effectively distribute the capabilities offered by robots: service-oriented robots [9].

This paper proposes a formal framework for coordinated simulation of heterogeneous service-oriented applications. It relies on the SCA-ASM language [32] that combines the OSOA open standard model Service Component Architecture (SCA) [29] for heterogeneous service assembly in a technology agnostic way, with the formal method Abstract State Machine (ASM) [12] able to model notions of service behavior, interactions, and orchestration [10, 7, 11] in an abstract but executable way. A designer may use the proposed framework to provide abstract implementations in SCA-ASM of (i) mock components (possibly not yet implemented in code or available as off-the-shelf) or of (ii) core components containing the main service composition or process that coordinates the execution of other components (possibly implemented using different technologies) providing the real computation. He/she can then validate the behavior of the overall assembled application, by configuring these SCA-ASM models in place within an SCA-compliant runtime platform as implementation of (mock or core) components and then execute them together with the other (local or remote) components implementations according to the chosen SCA assembly.

We, in particular, show the usage of our framework through a Robotics Task Coordination scenario from a case study [27] of the EU project BRICS [14]. In Robotics, service-oriented components embed the control logic of the application. They cooperate with each other locally or remotely through a communication network to achieve a common goal and compete for the use of shared resources, such as a robot sensors and actuators, the robot functionality, and the processing and communication resources. Cooperation and competition are forms of interactions among concurrent activities. So, in this domain, applications are very workflow-oriented and require developing coordination models explicitly [16].
ASMs provide a general method to combine specifications on any desired level of abstraction, ground modeling (requirements capture) techniques and stepwise refinement to executable code providing the basis for experimental validation and mathematical verification [12]. ASM rigorousness, expressiveness, and executability allow for the definition and analysis of complex structured services interaction protocols in a formal way but without overkill. Moreover, the ASM design method is supported by several tools [22, 5], useful for validation and verification of ASM-based models of services.

This paper is organized as follows. Section 2 provides background on SCA and ASMs. Section 3 presents the Robotics Task Coordination case study that will be used throughout the paper. Section 4 describes the proposed framework for coordinated simulation of service-oriented applications. Section 5 describes some related works, while Section 6 reports our lesson learned in developing the case study. Finally, Section 7 concludes the paper and sketches some future work.

2 Background on SCA and ASMs

**Service Component Architecture** SCA is an XML-based metadata model that describes the relationships and the deployment of services independently from SOA platforms and middleware programming APIs (as Java, C++, Spring, PHP, BPEL, Web services, etc.). SCA is supported by a graphical notation (a metamodel-based language developed with the Eclipse-EMF) and runtime environments (like Apache Tuscany and FRAscaTI) that enable to create service components, assemble them into a composite application, provide an implementation for them, and then run/debug the resulting composite application.

Fig. 1 shows an *SCA composite* (or *SCA assembly*) as a collection of SCA components. Following the principles of SOA, loosely coupled service components are used as atomic units or building blocks to build an application.

An *SCA component* is a piece of software that has been configured to provide its business functions (operations) for interaction with the outside world. This interaction is accomplished through: *services* that are externally visible functions provided by the component; *references* (functions required by the component).

![An SCA composite (adapted from the SCA Assembly Model V1.00 spec.)](image)
wired to services provided by other components; properties allowing for the configuration of a component implementation with externally set data values; and bindings that specify access mechanisms used by services and references according to some technology/protocol (e.g. WSDL binding to consume/expose web services, JMS binding to receive/send Java Message Service, etc.). Services and references are typed by interfaces. An interface describes a set of related operations (or business functions) which as a whole make up the service offered or required by a component.

The provider may respond to the requester client of an operation invocation with zero or more messages. These messages may be returned synchronously or asynchronously.

Assemblies of service components deployed together are supported in terms of composite components consisting of: properties, services, service implementations organized as sub-components, required services as references, and wires connecting sub-components.

**Abstract State Machines** ASMs are an extension of FSMs [12] where unstructured control states are replaced by states comprising arbitrary complex data. The states of an ASM are multi-sorted first-order structures, i.e. domains of objects with functions and predicates (boolean functions) defined on them. The transition relation is specified by rules describing how functions change from one state to the next. There is a limited but powerful set of ASM rule constructors, but the basic transition rule has the form of guarded update “if Condition then Updates” where Updates is a set of function updates of the form $f(t_1, \ldots, t_n) := t$ which are simultaneously executed\(^6\) when Condition is true.

*Dynamic functions* are those changing as a consequence of agent actions (or updates). They are classified as: monitored (only read, as events provided by the environment), controlled (read and write), shared (read and write by an agent and by the environment or by another agent) and out (only write) functions.

Distributed computation can be modeled by means of multi-agent ASMs: multiple agents interact in parallel in a synchronous/asynchronous way. Each agent’s behavior is specified by a basic ASM. The predefined variable (or 0-ary function) self can occur in the model and is interpreted by each agent as itself.

Besides ASMs comes with a rigorous mathematical foundation [12], ASMs can be read as pseudocode on arbitrary data structures, and can be defined as the tuple (header, body, main rule, initialization): header contains the signature\(^7\) (i.e. domain, function and predicate declarations); body consists of domain and function definitions, state invariants declarations, and transition rules; main rule represents the starting point of the machine program (i.e. it calls all the other ASM transition rules defined in the body); initialization defines initial values for domains and functions declared in the signature.

\(\text{\textsuperscript{6}}\) \(f\) is an arbitrary \(n\)-ary function and \(t_1, \ldots, t_n, t\) are first-order terms. To fire this rule in a state \(S_i\), \(i \geq 0\), evaluate all terms \(t_1, \ldots, t_n, t\) at \(S_i\) and update the function \(f\) to \(t\) on parameters \(t_1, \ldots, t_n\). This produces another state \(S_{i+1}\) which differs from \(S_i\) only in the new interpretation of the function \(f\).

\(\text{\textsuperscript{7}}\) Import and export clauses can be also specified for modularization.
Executing an ASM $M$ means executing its main rule starting from a specified initial state. A computation $M$ is a finite or infinite sequence $S_0, S_1, \ldots, S_n, \ldots$ of states of $M$, where $S_0$ is an initial state and each $S_{n+1}$ is obtained from $S_n$ by firing simultaneously all of the transition rules which are enabled in $S_n$.

A lightweight notion of module is also supported. An ASM module is an ASM (header, body) without a main rule, without a characterization of the set of initial states, and the body may have no rule declarations.

An open framework, the ASMETA tool set [5], based on the Eclipse/EMF platform and developed around the ASM Metamodel, is also available for editing, exchanging, simulating, testing, and model checking models. Asmetal is the textual notation to write ASM models within the ASMETA tool-set.

The SCA-ASM modeling language By adopting a suitable subset of the SCA standard for modeling service-oriented components assemblies and exploiting the notion of distributed multi-agent ASMs, the SCA-ASM modeling language [32] complements the SCA component model with the ASM model of computation to provide ASM-based formal and executable description of the services internal behavior, services orchestration and interactions. According to this implementation type, a service-oriented component is an ASM endowed with (at least) one agent (a business partner or role instance) able to be engaged in conversational interactions with other agents by providing and requiring services to/from other service-oriented components’ agents. The service behaviors encapsulated in an SCA-ASM component are captured by ASM transition rules.

The ASM rule constructors and predefined ASM rules (i.e. named ASM rules made available as model library) used as basic SCA-ASM behavioral primitives are recalled in Table 1 by separating them according to the separation of concerns computation, communication and coordination. In particular, communication primitives provide both synchronous and asynchronous interaction styles (corresponding, respectively, to the request-response and one-way interaction patterns of the SCA standard). Communication relies on a dynamic domain Message that represents message instances managed by an abstract message-passing mechanism: components communicate over wires according to the semantics of the communication commands reported above and a message encapsulates information about the partner link and the referenced service name and data transferred. We abstract, therefore, from the SCA notion of binding.

Fault/compensation handling is also supported (see [32]), but their SCA-ASM constructs are not reported here since they are not used in the case study.

Indeed, we adopt the default SCA binding (binding.sca) for message delivering, i.e. the SOAP/HTTP or the Java method invocations (via a Java proxy) depending if the invoked services are remote or local, respectively.
Table 1. SCA-ASM rule constructors for computation, coordination, communication

<table>
<thead>
<tr>
<th>COMPUTATION AND COORDINATION</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Skip rule</strong></td>
</tr>
<tr>
<td><strong>Update rule</strong></td>
</tr>
<tr>
<td><strong>Call rule</strong></td>
</tr>
<tr>
<td><strong>Let rule</strong></td>
</tr>
<tr>
<td><strong>Conditional rule</strong></td>
</tr>
<tr>
<td><strong>Iterate rule</strong></td>
</tr>
<tr>
<td><strong>Parallel rule</strong></td>
</tr>
<tr>
<td><strong>Forall rule</strong></td>
</tr>
<tr>
<td><strong>Choose rule</strong></td>
</tr>
<tr>
<td><strong>Split rule</strong></td>
</tr>
<tr>
<td><strong>Spawn rule</strong></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>COMMUNICATION</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Send rule</strong></td>
</tr>
<tr>
<td><strong>Receive rule</strong></td>
</tr>
<tr>
<td><strong>SendReceive rule</strong></td>
</tr>
<tr>
<td><strong>Reply rule</strong></td>
</tr>
</tbody>
</table>

3 Running case study: a robotics tasks coordination

We propose a simple scenario where a laser scanner offers its scan service to different clients, which compete for the use of this shared resource. The scenario is defined by three participants:

- A **Laser Scanner**, which executes scans of the environment on demand and writes the acquired values on a data buffer. A scan is a sequence of measures executed in a single task (for example 360 values, one for each degree). We suppose that the Laser Scanner allows its client to request a scan from an initial angle (start) to a finale one (end) defined as the number of steps between start and end.

- A **3D Perception application**, which requests the measures to the Laser Scanner in order to generate a set of meshes that describe the surface of the objects present in the environment.

- An **Obstacle Avoidance application**, which requests the measures to the Laser Scanner in order to detect the obstacles along the robot path.

The proposed scenario is subjected to the following requirements:
1. The laser scan activity requires a certain amount of time to be completed. This time is not fixed, and depends on the number of measures requested by the client. During this time the client could have the need of executing other activities and so it does not have to wait for the scan termination.

2. A client could request a single scan or multiple scans (for example 4 scans composed each one by 20 measures).

3. While the Laser Scanner is executing a scan requested by a client A, a client B could require another scan. These requests have to be managed according to one of the following policies:
   – Policy 1: Discard the scan request.
   – Policy 2: Queue the scan request.

Moreover, it is assumed that different clients could simultaneously access to the services offered by the Laser Scanner and that client requests are asynchronous, i.e. the client requests a scan to the Laser Scanner and then it continues to execute its work. In this case the interactions between the clients and the Laser Scanner have to be managed by a third part: a coordinator. This coordinator, Sensor Coordinator, is in charge of forwarding the clients requests to the Laser Scanner and so it has to manage the concurrent access of the clients.

**High-level solution** In order to keep the example simple to expose, we assume in this paper\(^9\) to address only the request management policy 1, i.e. if a request is received while the laser is already scanning the new request will be discarded. With this assumption, the Sensor Coordinator behavior can be captured, as first high-level model, by the finite state machine shown in Fig. 2.

Essentially, the Sensor Coordinator receives a request of one or \(n\) scans from a client. According to the followed policy (see above) the new request could be discarded, or queued or forwarded (the normal case) to the Laser Scanner. When the request is forwarded, the Laser Scanner starts the scanning work and sends a notification (\textit{Ack}) to the Sensor Coordinator in order to inform it that the scan has started. Depending on the number of scan requested, the Sensor Coordinator will forward to the Laser Scanner one or more single scans. In case of multiple scans, the Sensor Coordinator will forward \(n\) single scan requests to the Laser Scanner (to this purpose, the count variable \(\text{remScans}\), initially set to \(n\), is used and decremented at each forward). The Laser Scanner then writes each measure on the Measures Buffer until the final angle is reached, and it finally sends a notification (\textit{Done}) to the Sensor Coordinator in order to inform it that the scan is finished. At this point, if there are not remaining scans to execute (\(\text{remScans}\) is equal to 0) it sends a notification to the client in order to inform it that the new measures are available on the Buffer. The client then can access the Measures Buffer to read the measures.

\(^9\) Details on different variants of this scenario can be found in [27].
The application is heterogeneous: by the icons attached to components, the Sensor Coordinator is implemented in ASM, while the other two components in Java. The clients are considered external entities interacting with the Sensor Coordinator and with the Measures Buffer through the services offered (promoted) by the composite. More precisely, a client could request a scan by means of the service `SensorCoordinating` and could access the Measures Buffer by means of the service `MeasuresBufferReading`.

The definition of the service interfaces is reported in the listing 1.1 using the Java interface construct as IDL (Interface Definition Language). Note that, the interface `EventObserving` is implemented by the Sensor Coordinator to manage the notification received from the Laser Scanner.\textsuperscript{10}

The ASM (abstract) implementation of the SCA Sensor coordinator’s behavior will be provided later in Sect. 4.1. For the sake of space, the Java implementation code of the other components is not reported.

**Listing 1.1. Service interfaces definition in Java**

```java
public interface MeasuresBufferReading {
    public LaserScan getScan();
}

public interface MeasuresBufferWriting {
    public void writeMeasure(LaserMeasure measure);
}

public interface LaserScanning {
    /\*\* @param from: point from which the laser starts the scan
       * @param numOfSteps: number of steps of the scan */
    @OneWay public void scan(int from, int numOfSteps);
}

public interface SensorCoordinating {
    /\*\* @param from: point from which the laser starts the scan
       * @param numOfSteps: number of steps of the scan
       * @param numOfScans: number of scans required */
    @OneWay public void request(int from, int numOfSteps, int numOfScans);
}

public interface EventObserving {
    /\*\* @param event: it describe the type of event.
       * For the laser scanner valid values are "Ack" and "Done"
       */
    public void update(String event);
}
```

\textsuperscript{10} So far it is used as a service to resemble a callback (not yet supported in SCA-ASM).
4 Coordinated simulation framework

The proposed framework relies on the SCA-ASM language originally presented in [32] as a formal and abstract component implementation type to cover computation, communication, and coordination aspects during early execution (or simulation) of an SCA assembly of a heterogeneous service-oriented application. ASMs can be adopted to provide abstract implementations (or prototypes) of mock components, or to implement “core” components that contain the main service composition or coordination process that guides the application’s execution. The framework relies also on other SCA component implementation types (such as Java, Spring, C++, etc., see [29]) to include components providing the real computation services used by the core component(s) and these components can themselves require services provided by other local or remote components.

The framework was developed by integrating the Eclipse-based SCA Composite Designer, the SCA runtime platform Tuscany [35], and the simulator AsmetaS of the ASM toolset ASMETA [5]. This environment11 allows us to graphically model, compose, analyze, deploy, and execute heterogeneous service-oriented applications in a technologically agnostic way. As described and exemplified below, an heterogeneous SCA assembly (or composition) of service-oriented components (implemented in ASM or in another implementation language) can be graphically produced using the SCA Composite Designer and also stored or exchanged in terms of an XML-based configuration file. This last file is then used by the SCA runtime to instantiate and execute the system by instrumenting AsmetaS and other execution infrastructures in an unique environment (see Fig. 5).

4.1 Service component implementation and configuration

Through the considered case study, we here show the use of the ASM implementation type (i.e. of the SCA-ASM language) for SCA components. This implementation type provides a behavioral specification that is essential for performing early prototyping, functional simulation and formal verification to prove (verifiable) correctness and reliability of both single services and the overall assembly.

Service component implementation The following listings report the ASM (abstract) implementation of the Sensor Coordinator component (request management policy 1). To this purpose, the AsmetaL textual notation to write ASM models within the ASMETA tool-set is used. Two grammatical conventions must be recalled: a variable identifier starts with $; a rule identifier begins with “r_”.

Listing 1.2 shows the header of the ASM. The import clauses include the ASM modules of the provided service interfaces (SensorCoordinating and EventObserving) and required interfaces (the LaserScanning interface) of the component, annotated, respectively, with @Provided and @Required. The @MainService annotation, when importing the SensorCoordinating interface, denotes the main service (read: main component’s agent) that is responsible for initializing the

component’s state (in the predefined \texttt{r\_init} rule). The signature of the machine contains declarations for: references (shared functions annotated with \texttt{@Reference}), which are abstract access endpoints to services, back references to requester agents (shared functions annotated with \texttt{@Backref}), and declarations of ASM domains and functions, which are used by the component for internal computation only. In particular, the variable (a controlled 0-ary function) \texttt{ctl\_state} stores the current control state of the ASM.

<table>
<thead>
<tr>
<th>Listing 1.2. ASM header of the Sensor Coordinator component</th>
</tr>
</thead>
<tbody>
<tr>
<td>module SensorCoordinator import STDL/StandardLibrary import STDL/CommonBehavior @MainService import SensorCoordinating @Provided import EventObserving @Required import LaserScanning export * signature: @Reference shared laserScanning : Agent -&gt; LaserScanning @Backref shared clientSensorCoordinating : Agent -&gt; Agent @Backref shared clientEventObserving : Agent -&gt; Agent enum domain State = {IDLE</td>
</tr>
</tbody>
</table>

The body of the ASM (see Listing 1.3) includes definitions of the services (transition rules annotated with \texttt{@Service}) \texttt{r\_request} and \texttt{r\_update}, the main transition rule \texttt{r\_SensorCoordinator} (that takes by convention the same name of the component), the transition rule with the predefined name \texttt{r\_init} that is invoked in the initialization to set up the internal component state (i.e. values of controlled functions), and another utility rule named \texttt{r\_acceptRequest}.

The \texttt{r\_request} service is in charge of requesting a scan to the laser scanner. When the rule is called, it executes in parallel the following actions: sets the state of the ASM to \texttt{BUSY}; stores the arguments of the requested scan, invokes (by a send action) the service \texttt{scan}, which is provided by the service Laser Scanning.

The \texttt{r\_update} service is in charge of receiving the notification from the laser scanner and updating the control state by resembling the FSM shown in Fig. 2.

The \texttt{r\_acceptRequest} rule advances the control state of the machine properly according to the arriving service requests (the input parameter \texttt{$r$}). In case of a new scan request (\texttt{r\_request}), this is removed from the requests queue (by the \texttt{r\_wreceive} action) and the input is stored in the variable \texttt{paramScan}. A direct invocation of the \texttt{r\_request} service then follows if the input is defined. In case, instead, of a notification (\texttt{r\_update}) from the laser scanner, the request is removed from the requests queue (by the \texttt{r\_wreceive} action) and in case the
input (stored in the variable `event`) is defined the service `r_update` is invoked. Note that all the scan requests received while the scanner is already scanning are discarded (what the policy 1 defines).

Listing 1.3. ASM body of the Sensor Coordinator component

```asm
//State invariant: Number of scans required by a client must be non negative
invariant inv_neverNeg over remScans(): not(remScans < 0)
//@Service
rule r_request($a in Agent,$from in Integer,$steps in Integer,$nScans in Integer)=
  par
  ctl_state($a) := BUSY
  from($a) := $from
  steps($a) := $steps
  remScans($a) := $nScans - 1
  r_wsend[laserScanning($a),"r_scan(Agent,Integer,Integer)",($from,$steps)]
endpar
//@Service
rule r_update($a in Agent,$event in String) =
  if (ctl_state($a)=BUSY and $event="Ack")
  then
    ctl_state($a) := SCANNING
  else if (ctl_state($a)=SCANNING and $event="Done" and remScans($a)>0)
  then
    par
      ctl_state($a) := BUSY
      remScans($a) := remScans($a) - 1
      r_wsend[laserScanning($a),"r_scan(Agent,Integer,Integer)",(from($a),steps($a))]
  endpar
  else if (ctl_state($a)=SCANNING and $event="Done" and remScans($a)=0)
  then
    ctl_state($a) := IDLE
  endif endif
//@Service
rule r_acceptRequest ($a in Agent,$r in String) =
  if (ctl_state($a)=IDLE and $r="r_request(Agent,Integer,Integer,Integer)")
  then
    seq
      r_wreceive[clientSensorCoordinating($a),"r_request(Agent,Integer,Integer,Integer)",paramScan($a)]
      if (isDef(paramScan($a)))
      then
        r_request[$a,first(paramScan($a)),second(paramScan($a)),third(paramScan($a))]
      endif
e ndseq
  else if (not ctl_state($a)=IDLE and $r="r_update(Agent,String")
  then
    seq
      r_wreceive[clientEventObserving($a),"r_update(Agent,String)",event($a)]
      if (isDef(event($a)))
      then
        r_update[self,event($a)]
      endif
e ndseq
e ndif
//@Service
rule r_init($a in SensorCoordinating) = //for the startup of the component
  par
    status($a) := READY
    ctl_state($a) := IDLE
    from($a) := 0
    steps($a) := 0
    remScans($a) := 0
endpar
```
The \texttt{r_SensorCoordinator} rule is the program of the main component’s agent and is invoked every times a client requests a service offered by the Sensor Coordinator. This rule simply forwards the request to the \texttt{r_acceptRequest} rule.

Finally, the \texttt{r_init} rule is called during initialization of the component execution. This rule simply sets the status of the agent to \texttt{READY}, the control state to \texttt{IDLE} and initializes the scan parameters to 0.

The ASM definitions of the sensor coordinator’s provided interfaces are reported in the listing 1.4 using the AsmetaL notation. They are ASM modules containing only declarations of business agent types (subdomains of the pre-defined ASM domain \texttt{Agent}), and of business functions (ASM out functions).

\textbf{Listing 1.4. ASM definition of the Sensor Coordinating interface}

\begin{verbatim}
//@Remotable
module SensorCoordinating
import STDL/StandardLibrary
import STDL/CommonBehavior
export *
signature:
// the domain defines the type of this agent
domain SensorCoordinating subsetof Agent
// out is a function that implements the provided service
out request: Prod(Agent,Integer,Integer,Integer) => Rule
definitions:
//@Remotable
module EventObserving
import STDL/StandardLibrary
import STDL/CommonBehavior
export *
signature:
domain EventObserving subsetof Agent
out update: Prod(Agent,String) => Rule
definitions:
\end{verbatim}

\textbf{Listing 1.5. XML configuration file}

\begin{verbatim}
<?xml version="1.0" encoding="UTF-8" standalone="no"?>
<sca:composite xmlns:sca="http://www.osoa.org/xmlns/sca/1.0" xmlns:asm="http://asm"
name="Sensor" targetNamespace="http://eclipse.org/CaseStudy/src/Sensor">
  ...
  <sca:component name="SensorCoordinator">
    <asm:implementation.asm location="SensorCoordinator.asm"/>
    <sca:reference name="laserScanning"/>
    <asm:interface.asm location="SensorCoordinating.asm"/>
  </sca:service>
  <sca:component>
    ...
  </sca:component>
  ...
</sca:composite>
\end{verbatim}

\textbf{Service component configuration} Component metadata, describing which services are required and provided by a component, and information that allow the SCA runtime to locate (locally or remotely) the component implementation, must be provided in the SCA XML composite file. Listing 1.5 shows a fragment of the SCA XML composite file regarding the metadata of the Sensor Coordinator component that is implemented (by the tag \texttt{implementation.asm}) in ASM.

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4.2 In-place simulation of SCA-ASM models

SCA-ASM components use annotations to denote services, references, properties, etc. With this information, as better described below, an SCA runtime platform (Tuscany in our case) can create a composition (an application) by tracking service references (i.e. required services) at runtime and injecting required services into the component when they become available.

![Fig. 4. Instantiating and invoking ASM implementation instances within Tuscany](image)

**In-place ASM execution mechanism.** Fig. 4 illustrates how the ASM implementation provider\(^\text{12}\) sets up the environment (the container) within Tuscany for instantiating and handle incoming/outgoing service requests to/from an ASM component implementation instance (like component A in the figure) by instrumenting the ASM simulator AsmetaS. Currently, the implementation scope of an SCA-ASM component is composite, i.e. a single component instance – a single main ASM instance (see the main ASM for component A in Fig. 4) – is created within AsmetaS for all service calls of the component\(^\text{13}\). This main ASM is automatically created during the setting up of the connections and it is responsible for instantiating the component agent and related resources, and for listening for service requests incoming from the protocol layer and forward them to the component agent instance (see component A in Fig. 4). Executing an ASM component implementation means executing its main ASM. For each reference, another entity (i.e. another ASM module) is automatically created (and instantiated as ASM agent within the main ASM of the component) as “proxy” for a remote component (see the ASM proxy for component B in Fig. 4) for making an outbound service call from the component. Using a terminology adopted in

\(^{12}\) The Tuscany core delegates the start/stop of component implementation instances and related resources, and the service/reference invocations, to specific implementation providers that typically respond to these life-cycle events.

\(^{13}\) We postpone as future work the implementation of the other two SCA implementation scopes, *stateless* (to create a new component instance on each service call) and *conversation* (to create a component instance for each conversation).
the Java Remote Method Invocation (RMI) API, this proxy ASM plays the role of stub to forward a service invocation (and their associated arguments) to an external component’s agent, and to send back (through the ASM rule \texttt{r.replay}) the result (if any) to the invoker component’s agent (the agent of the component \texttt{A} in Fig. 4). The main ASM, instead, plays the role of skeleton, i.e. a proxy for a remote entity that runs on the provider and forward (through the ASM rule \texttt{r.sendreceive}) client’s remote service requests (and their associated arguments) to the appropriate component’s agent (usually the main agent of the component), and then the result (if any) of the invoked service is returned to the client component’s agent (via stubs). For the sake of space, the ASM implementation of the stub and skeleton (as generated by the runtime) for the component Sensor Coordinator is not reported.

When an ASM implementation component is instantiated, the Tuscany runtime also creates a value for each (if any) externally settable property (i.e. ASM monitored functions, or shared functions when promoted as a composite property, annotated with \texttt{@Property}). Such values or proxies are then injected into the component implementation instance. A data binding mechanism also guarantees a matching between ASM data types and Java data types, including structured data, since we assume the Java interface as IDL for SCA interfaces.

Fig. 5 shows a simulation snapshot of the considered case study where the Sensor Coordinator changes state from \texttt{IDLE} to \texttt{BUSY} (see also the rule \texttt{r.request} in the Listing 1.3) after receiving a first scan request from a client.
Other ASM execution features Useful features are currently supported by the AsmetaS simulator when running within the SCA Tuscany platform.

State invariant checker: AsmetaS implements an invariant checker, which at the end of each transition execution checks if the invariants (if any) expressed over the state of the currently executed SCA-ASM component are satisfied or not. If an invariant is not satisfied, AsmetaS throws an InvalidInvariant-Exception, which keeps track of the violated invariant. Listing 1.3 shows an example of state invariant (inv_neverNeg) for the Sensor Coordinator. It states that the number of scans required by a client must be non negative.

Consistent Updates checking: The simulator also includes a checker for revealing inconsistent updates. In case of inconsistent updates an Update Clash Exception is thrown by reporting the location which is being inconsistently updated and the two different values which are assigned to that location. The user, analyzing this error, can detect the fault in the ASM component implementation.

Logging: AsmetaS produces a minimal output to show the current state and the update set. The user can also inspect how the simulator performs particular tasks (including the evaluation of terms, the building of update set for rules, and the variables substitution) by providing a log4j configuration file.

5 Related work

Some works devoted to provide software developers with formal methods and techniques tailored to the service domain also exist (see, e.g., the survey in [8] for the service composition problem), mostly developed within the EU projects SENSORIA [33] and S-Cube [28]. Several process calculi for the specification of SOA systems have been designed (see, e.g., [23, 25, 17]). They provide linguistic primitives supported by mathematical semantics, and verification techniques for qualitative and quantitative properties [33]. Still within the SENSORIA project, a declarative modeling language for service-oriented systems, named SRML [34], has been developed. SRML supports qualitative and quantitative analysis techniques using the UMC model checker [1] and the PEPA stochastic analyzer 14.

Compared to the formal notations mentioned above, the ASM method has the advantage to be executable. On the formalization of the SCA component model, some previous works, like [19, 20] to name a few, exist. However, they do not rely on a practical and executable formal method like ASMs. In [26], an analysis tool, Wombat, for SCA applications is presented; this approach is similar to ours as the tool is used for simulation and verification tasks by transforming SCA modules into composed Petri nets. There is not proven evidence, however, that this methodology scales effectively to large systems.

An abstract service-oriented component model, named Kmelia, is formally defined in [6, 3] and is supported by a prototype tool (COSTO). In the Kmelia model, services are used as composition units and services behavior is modeled by labeled transition systems. Our proposal is similar to the Kmelia approach; 14 http://www.dcs.ed.ac.uk/pepa/
however, we have the advantage of having integrated our SCA-ASM component model and the ASM-related tools with an SCA runtime platform for a more practical use and an easier adoption by developers.

Within the ASM community, the ASMs have been used for the purpose of formalizing business process notations and middleware technologies related to web services, such as [13, 10, 21, 2] to name a few. Some of these previous formalization efforts, as explained in [32], are at the basis of our work.

Concerning the Robotics domain, [24] proposes a new approach for coordinating the behavior of Orocos RTT (Open Robot Control Software Real Time Toolkit) [31] components. Orocos RTT is a C++ framework, which allows the design and the deployment of component-based robotics control systems. The proposed approach defines the behavior of single components and of entire systems by means of a minimal variant of the UML hierarchical state-charts, which is called reduced FSM (rFSM). The main advantages of the rFSMs are their hierarchical composability and their applicability in hard-real time applications. Furthermore, despite they are currently used only with Orocos, rFSMs are totally framework independent. The main differences between the ASMs and the rFSMs are that rFSMs do not allow the execution of parallel agent actions and parallel states; moreover, they do not have the universality and broad application of the ASMs, and do not offer the same flexibility and tools provided by the ASMs.

6 Lesson learned

We have shown how formal high-level ASM models of service-oriented components can be assembled together with real components through the SCA framework and how we manage the coordination of the overall resulting application by means of the ASM formalism for prototyping and simulation purposes. We experienced that the use of two different frameworks for modeling two different concerns (SCA and its various implementation types for computation, and ASM for coordination) improves the level of flexibility and reusability.

We have shown this by means of a use case in the Robotics field, where flexibility and reusability are very challenging issues [14–16]. In general, robotics software require and provide a number of different functionalities, which are typically encapsulated in components that cooperate and compete in order to control the behavior of a robot. Cooperation and competition are forms of interaction among concurrent activities and so they have to be coordinated. In order to achieve a good level of reusability and flexibility the coordination and the computation (how the component provides the service) need to be managed separately. So by our experience, the service paradigm seems promising also in the Robotics domain. In particular, we appreciated the possibility to change the coordination policies (see [27]) without modifying the implementation of the services provided by components merely dedicated to computation (such as sophisticated algorithms), thus improving the level of flexibility and reusability.
7 Conclusion and future directions

We presented a practical framework for early service design and prototyping that combines the SCA open standard and the ASM formal support to assemble service-oriented components as well as intra- and inter-service behavior. The framework is supported by a tool that exploits the SCA runtime Tuscany and the toolset ASMETA for model execution and functional analysis. The effectiveness of our framework was experimented through various case studies of different complexity and heterogeneity. These include examples taken from the SCA Tuscany distribution, the case study of the EU project BRICS [14] presented here, and also a scenario of the Finance case study of the EU project SENSORIA [33].

We plan to support more useful SCA concepts, such as the SCA callback interface for bidirectional services and an event-based style of interaction. We want also to enrich the SCA-ASM language with interaction and workflow patterns based on the BPMN specification. We also plan to support pre/post-conditions defined on services for contract correctness checking in component assemblies.

On the functional analysis side, we plan to experiment the framework on large distributed application scenarios involving SCA assemblies deployed on various local and remote hosting nodes. We want also to try the use of SCA-ASM models as oracles for reasoning and testing about real components implementations, including but not limited to, conformance testing and run-time monitoring.

References


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Verifying Temporal Properties of Use-Cases in Natural Language

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Abstract. This paper presents a semi-automated method that helps iteratively write use-cases in natural language and verify consistency of behavior encoded within them. In particular, this is beneficial when the use-cases are created simultaneously by multiple developers. The proposed method allows verifying the consistency of textual use-case specification by employing annotations in use-case steps that are transformed into LTL formulae and verified within a formal behavior model. A supporting tool for plain English use-case analysis is currently being enhanced by integrating the verification algorithm proposed in the paper.

Keywords: Use-Cases, Behavior Modeling, Verification, Natural Language, Label Transition System, Model-Checking, Requirements Engineering

1 Introduction

In typical software development practice, majority of the requirement documents created in the early phase of a project, are written in natural language [10]. Such a specification is therefore inherently imprecise, ambiguous, and a potential source of contradictions. An important issue is that in a large software project, the specification phase involves collaboration among a number of team members who express their personal views in natural language. In such an environment, there is a high chance of conflicts among individual parts of the specification.

Use-cases are traditionally used in requirement specification because they can easily capture the behavior of a system under discussion (SuD) from the perspective of different actors. Usually, SuD may be equalled to a component where a use-case describes a part of the interaction between the component and its environment.

Since the inclusion of use-cases into the UML standard [14], their use has been greatly extended, making them a mandatory requirement for any object-oriented software development project. As stressed by Cockburn [3] and Larman [9], the main asset

\footnote{For example, the Agile software development methodology proposes teams of 5-9 people.}
of use-cases is that behavior is encoded in natural language and thus accessible to a wide range of stakeholders of a project.

Although an isolated use-case can clearly describe a simple scenario, the overall behavior of combined use-cases may become quite blurry. In particular, the problem can easily appear in specifications where use-cases are composed using \textit{include} and \textit{precede} relationships [3].

The intended behavior expressed by use-cases contains implicit temporal dependencies that are likely violated during the iterative development. Because late detection of such errors leads to significantly higher costs of a project [2], writers of the specification greatly benefit from tools that help them keep the textual specification consistent and that warn them about potential violations immediately during writing.

Motivation example: Figure 1 shows a pair of the dependent use-cases $U_1$ and $U_2$ specified as a sequence of English sentences ($U_1$ includes $U_2$). The final text of these use-cases was created in 3 iterations. In the first iteration, an initial version was created with just a simple success scenario. In the second iteration, the use-case $U_2$ was refined by introducing an optional branch (variation) aborting $U_2$. However, such specification is not consistent: $U_1$ does not consider a possible abort in $U_2$. In more detail, there is a possible trace leading to usage of an unavailable item (in case of the aborted $U_2$, GPM in $U_1$ is unavailable – Figure 1).

Problem statement. Such an inconsistency may be detected only when both use-cases are put into context of one another using the \textit{include} relationship. This makes such inconsistencies difficult to notice, especially when specification is large with many use-cases and includes relationships.
So it would be desirable to propose a method that, in an automatized way, detects such an inconsistency and issues a warning. In the example, as a reaction to such a warning, $U_1$ could be manually extended by adding an abort-handling branch to affect the set of traces that involve branching transitions. Verification would now succeed because the traces involving the abort step in $U_2$ would be limited to the abort-handling branch.

**Goal.** Thus, the goal of this paper is to present a method that allows an early detection of violation of temporal dependencies of use-case steps. The proposed method (Use-Case Temporal Verification – UCTV) allows automated derivation of a formal behavior model (LTS) from use-cases in plain English. Moreover, by adding annotations to use-case steps, it is possible to verify temporal properties in an automatized way in order to identify inconsistencies within the original specification. The detected errors are presented to the user as erroneous traces. For automated transformation of the use-cases into the formal model and verification of temporal dependencies, we designed a software tool (REPROTOOL), which stems from the PROCASOR tool [12,4,15] designed earlier in our group.

Other approaches exist that aim at extracting behavior models from text, for example authors of [19] describe how to generate UML Activity Diagrams from use-cases. The method uses restriction rules [20] imposed on the use-case step sentences. In [7], a method for deriving message sequence charts from textual scenarios is described.

Several languages and formalisms for behavior modeling of software systems have been proposed. They range from very generic ones (e.g., process algebras [6,13]), to those specific to components (e.g., Darwin [11], Interface automata [1], or Threaded Behavior Protocols [8]).

To achieve the goal, the paper is structured as follows: In Section 2 we overview the main concepts in UML use-cases as the terminology base used further in the paper. Section 3 describes how users interact with an application that implements our method. In Section 4 we explain the algorithm in detail, while Section 6 concludes the paper.

### 2 Use-cases in natural language and UML

The prevalent practice of capturing use-cases is to use textual notation and natural language. Further, UML Use-Case Diagrams provide means for establishing relations among use-cases.

Although there are different styles of writing use-cases, for our purposes we consider the format depicted in Figure 1 and 5. This format is taken from the book [3] as it is widely accepted.

With regard to the structure of a use-case, the *main success scenario* of a use-case consists of several steps that contribute to achieving the use-case goal. Alternative scenarios can be expressed using *variations* and *extensions*. The difference between extensions and variations is that a variation *replaces* the step to which it is attached, while an extension provides *optional branching* from its parent step. For illustration, consider the use-case $U_1$ in Figure 5. There is a variation $2a$ attached to the step 2 which means that 2 and 2a are mutually exclusive branches. On the other hand, the use-case $U_2$ con-
tains an extension 1b which means that the step 1 is always executed before the optional 1b branch.

2.1 Actions in use-case steps

It has been advised by practitioners, e.g. in [3], to use simple sentences when writing use-cases. A sentence should encode a single action, which is either (a) interaction between an actor and SuD, (b) internal action within SuD, or (c) special action (see below). As to the structure of a sentence, in English it should conform to the SVDPI pattern (Subject, Verb, Direct-Object, Preposition, and Indirect-Object); this is very important for an automated processing. The following special actions are introduced in the UCTV method:

Goto action: The trace advances by another step (indicated by this action) within the same use-case. This action is typically used to express looping. Example: Goto step 1. (See Figure 5 use-case U2 step 2a2).

Include action: Similar to calling a procedure, the trace advances in the included use-case, when it is finished, the include action is concluded. Example: Include use-case “Generate city” (e.g. use-case U2, step 1 in Figure 5).

Abort action: The use-case execution is aborted. However, if the aborted use-case U3 was included into another use-case U2, the trace immediately advances in U2. Example: Use-case aborts (e.g. use-case U3, step 2a1 in Figure 5).

2.2 Relations in the UML Use-Case Diagrams

UML provides means for expressing dependencies among use-cases using stereotyped relations in the UML Use-Case Diagrams. The UCTV method takes into account the «includes» (via the include special action) and «precedes» UML relationships:

U1 «includes» U2: The include relationship allows inserting the behavior from one use-case into another. It minimizes duplication and improves comprehension of the whole specification when used carefully. The use of include means that at the given point in use-case A, the trace advances over the steps in B and when B is finished, it returns back to A [9].

U1 «precedes» U2: Rosenberg and Stephens in the book [16] define the precedence relationship as: The use-case U1 must take place in its entirety before another use-case U2 even begins, i.e. there is temporal precedence in which U1 must occur before U2. For example a Login use-case must be completed before Checkout is begun.

We use the Prec precedence relation formed by the pairs of use-cases, in which first use-case precedes the second one.
3 User’s perspective

Before we present the UCTV in a formal way, let us describe use-case design from the user’s perspective. Figure 2 contains a screenshot from our application REPROTOOL that the user employs when writing use-case specification.

![Screenshot from the REPROTOOL application.](image)

In the first phase, the user creates several use-cases with steps written in English prose. Each step is automatically parsed and the sentence is transformed into a linguistic parse tree. Depending on the sentence structure, the type of the action is derived automatically or set manually by the user. This way, REPROTOOL derives LTS from the use-cases and renders a graphical representation of the LTS as depicted in Figure 2.

In the next phase, the user can assign annotations to individual steps to define precedence relations determining temporal dependencies among use-cases and their steps. These will be verified in the next phase.

When looking at the motivation example, the temporal dependency between $U_1$ and $U_2$ can be captured using a pair of annotations – use:item and create:item (for illustration see Figure 3 providing annotated use-cases and capturing their creation in iterations). The semantics of them is that in each trace containing a step with the use:item annotation, any other step with a create:item annotation has to precede the former (pairwise).

At some point, during the iterative process of writing use-cases, the user initiates the verification procedure performed within the REPROTOOL application.

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2 REPROTOOL is based on Eclipse and uses Eclipse Modeling Framework (EMF) as a tool for data representation. The application is still under development and not yet completely finished, see [http://code.google.com/a/eclipselabs.org/p/reprotool/](http://code.google.com/a/eclipselabs.org/p/reprotool/).
Fig. 3. Verification of dependent use-cases with aborts using temporal annotations

If a verification error is detected, the model-checker shows a trace that violates the temporal properties determined by annotations. After the verification is finished, the user can adjust the textual specification to fix the reported error by:

- Adding precedence relationships among use-cases, which fix the missing create annotation that the preceded use-case might have provided.
- Adding an abort-handling branch as seen in the motivation example (Fig. 3).
- Adjusting annotations of steps or rewriting/reorganizing them.

In the motivation example, after introducing the abort branch (variation 2a of $U_2$), the model-checker detected an error trace (Figure 3, Iteration #2). The user fixed the error by adding an extension 1a into $U_1$ (Figure 3, Iteration #3 and Figure 1, "Necessary correction").

In addition to the create-use annotation pair, we have also described the open-close annotation pair in the text below. These annotations cover the majority of dependencies among use-cases that we encountered in our survey [17]. However, since the UCTV method internally uses LTL formulae to capture desired temporal properties, our approach is also applicable for other annotations, the semantics of which can be described by LTL.

4 Verification of use-cases

In this section, we describe all the annotations and the REPROTOOL verification algorithm in detail.
4.1 Annotations in use-case steps

There are two types of annotations: (a) annotations expressing temporal dependencies (technically translated to LTL), and (b) annotations constraining the set of traces to be inspected by the model-checker.

(a) Annotations expressing temporal dependencies:

The create-use annotation pair: In all traces it must hold that for any step annotated by use:x there must previously appear a step annotated by create:x (as above, x is a user-chosen identifier). That is, if x is used, it must be created before. Next, it must hold that for each step annotated by create:x, there must be a trace reaching this step and then eventually reaching another step annotated by use:x. In other words, if x is created then it must be somewhere used in the future. An example is shown in Figure 5.

The open-close annotation pair: For any trace that with a step annotated by open:x, there must eventually appear a step annotated by close:x. Obviously, another open:x step is not allowed in between. In a similar vein, close:x cannot appear without a preceding open:x.

(b) Annotations constraining the traces:

The trace-on annotation pair: These annotations serve to control application of use-case variations.

Technically, the trace:x annotation marks with a flag x all the traces going through the step where this annotation appears. This flag may be later tested and used as a guard in branching via the annotation on:x. That is, a trace that goes through a step marked with trace:x annotation and reaches this branching state must continue using the step marked with an on:x annotation and a trace that does not go through a step marked with a trace:x annotation and reaches this branching state must continue using any other step going from this state.

Typically, this annotation pair is used when detecting unhandled aborts in use-cases. Figures 3 and 5 show examples.

4.2 The verification strategy

Verification of textual use-cases is done in two phases. First the precedences and includes are statically checked for presence of cyclic dependencies. Second, a dedicated type of LTS (use-case automaton) is built from the textual use-cases and model-checking is employed to verify temporal dependencies expressed by annotations. Figure 4 provides an overview of the verification phases and their steps.

Before the actual verification starts, the textual use-cases are parsed using the method described in [4] into an internal form where the sequence of steps, variations and extensions of steps, actions in use-case steps, and annotations of use-case steps are specifically represented. After the internal form has been created the verification proceeds as described below.

3 Strictly speaking, creating something without its usage is not an error. Nevertheless, since it is not a good practice, we consider such a trace to be erroneous.
Fig. 4. Verification algorithm as a pseudo-code implementation

Static check of precedences and includes In this phase, precedence and include use-case relationships are checked statically. The cyclic dependencies among use-cases represented in the internal form are detected by creating an oriented precedence and include graphs.

Model-checking of temporal dependencies In this phase, the internal form is used to build LTS-like structure (based on use-case automata). The annotations expressing temporal dependencies are converted into an LTL formula. Finally, a modified LTL model-checking algorithm (Use Case Model Checking – UCMC)\(^4\) is applied to verify the LTL properties. This phase comprises three steps:

- A use-case automaton with includes (UCAI) is built for each use-case. Basically, UCAI is an LTS with transitions corresponding to steps of a use-case. Specifically, it contains include transitions, which correspond to include steps in the use-case. Figure 5 shows an example of three textual use-cases and the corresponding UCAIs.
- By creating resolved use-case automata (RUCAs), includes in UCAIs are inlined. RUCA is obtained from UCAI by replacing each of the include transitions by inlining the reference automaton. See Figure 6a for an example of resolution of the automaton \(U_2\) from Figure 5. Moreover the annotations constraining the traces are converted into guards (controlled by dedicated variables) on the automata transitions (Figure 6c shows an automaton with guards).
- The annotations expressing temporal dependencies are converted to LTL formulae\(^5\). Figure 6b shows the automaton with annotations on transitions.

\(^4\)The model-checking algorithm cannot be used in its standard form since we consider also finite traces and LTS with guards – discussed in detail in Section 5.5
\(^5\)In this paper, we only consider LTL formulae corresponding to the create-use and open-close annotation pairs.
A set of automata which captures overall behavior of the system (OB) is created. Because the precedes relations define only a partial ordering of use-case applications, the overall behavior OB is determined by a set of all possible sequences of the use-case applications.

Technically, each such a sequence is represented by a RUCA created by a concatenations of the RUCAs representing the individual members of the sequence. Figure 6d shows an example of concatenation of the use-case automata U_1 and U_2 from Figure 5.

In the final step, the UCMC algorithm is used to verify each RUCA in OB against the extracted LTL formulae.

5 Theoretical background

In this section, we provide a formal definition of the key abstraction used by the UCTV method, specifically this includes UCAI, RUCA and a proof of the correctness of the UCMC algorithm.
Fig. 6. Visual representation of the construction of the verifying LTS: (a) Included LTS (green) is inlined to the base LTS (black), (b)+(c) annotations (blue) are initialized either as LTL variables or control variables with guards, (d) all final states of the preceded LTS (purple) are connected to the initial state of the base LTS (black). Note: These LTS automata correspond to use-cases from the Figure 5.

5.1 Use-case automaton with includes

We define use-case automaton with includes (UCAI) and the way it corresponds to a textual use-case. The correspondence is straightforward – steps of a textual use-case correspond to transitions of a use-case automaton with includes.

Definition 1 (Use-case automaton with includes–UCAI).

A use-case automaton with includes (UCAI) \( P = (V_P, V_P^{\text{init}}, V_P^{\text{abort}}, V_P^{\text{succ}}, A_P, \tau_P) \) consists of the following elements:

- \( V_P \) is a set of states
- \( V_P^{\text{init}} \subseteq V_P \) is a set of initial states. We require that \( V_P^{\text{init}} \) contains at most one state. If \( V_P^{\text{init}} = \emptyset \), then \( P \) is called empty.
- \( V_P^{\text{abort}} \subseteq V_P \) is a set of abort states.
- \( V_P^{\text{succ}} \subseteq V_P \) is a set of succeeded states. We require that \( V_P^{\text{succ}} \) contains at most one state.
- \( A_P = A_P^I \cup A_P^{\text{include}} \cup \{\epsilon\} \) is a set of all actions. \( A_P^I, A_P^{\text{include}} \) are mutually disjoint sets of internal and include actions, \( \epsilon \) is the empty action.
- \( \tau_P \subseteq V_P \times A_P \times V_P \) is a set of transitions.

Definition 2 (Annotation function). Let \( A_P \) be a set of actions of UCAI \( P \) and \( N \) a set of annotations. Annotation function \( A_f : \tau_P \mapsto 2^N \) maps a transition of \( P \) to a set of annotations.
Note that two instances of an annotation with an identical name – i.e. two on \(x\) annotations annotating different steps of use-case – are not considered as equal. Hence, there is no annotation that annotates two different steps.

**Definition 3 (Correspondence of a use-case to UCAI).** Let \(U\) be a use-case, let \(P = \langle V_P, V_P^{\text{init}}, V_P^{\text{abort}}, V_P^{\text{succ}}, A_P, \tau_P \rangle\) be UCAI. We say that \(P\) corresponds to \(U\) if for each step \(s_i\) of \(U\) there is the corresponding transition \(t_i = (s_i, a_i, s'_i) \in \tau_P\), \(s_i, s'_i \in V_P\), \(a_i \in A_P\) of \(P\) such that:

- If \(s_i\) is an include step then \(a_i \in A_P^{\text{include}}\), if \(s_i\) is an abort or a goto step \(a_i = \epsilon\), otherwise \(a_i \in A_P^{\epsilon}\).

\(V_P, V_P^{\text{init}}, V_P^{\text{abort}}, V_P^{\text{succ}}\), and \(\tau_P\) are defined as:

- If for any other step \(s_j \in U, s_j \neq s_i\) of \(U\) and corresponding transition \(t_j = (s_j, \alpha_j, s'_j)\) of the automaton is \(s'_i = s'_j\), then either \(s_j\) or \(s_i\) is a goto step, a last step of the variation, or a last step of an extension.

- If \(s_i\) is not the first step of the main success scenario, the first step of a variation, or the first step of an extension, let \(s_{i-1} \in U\) be a step preceding the step \(s_i\) and \((s_{i-1}, \alpha_{i-1}, s'_{i-1}) \in \tau_P\) a corresponding transition of \(P\). It holds that \(s'_{i-1} = s_i\).

- If \(s_i\) is the first step of the main success scenario of \(U\) then \(s_1 \in V_P^{\text{init}}\).

- If \(s_i\) is the last step of the main success scenario of \(U\) then \(s'_i \in V_P^{\text{succ}}\).

- If \(s_i\) is the first step of a variation of the step \(s_j \in U\) and let \((s_j, \alpha_j, s'_j) \in \tau_P\) be the transition of \(P\) corresponding to the step \(s_j\), then it holds that \(s_i = s_j\).

- If \(s_i\) is the first step of an extension of the step \(s_j \in U\) and let \((s_j, \alpha_j, s'_j) \in \tau_P\) be the transition of \(P\) corresponding to the step \(s_j\) and \(s_{j+1} \in U\) be the step following the step \(s_j\) and \((s_{j+1}, \alpha_{j+1}, s'_{j+1}) \in \tau_P\) corresponding transition, it holds that \(s_i = s'_{j+1}\).

- If \(s_i\) is the last step of a variation or an extension and it is an abort step, then \(a_i = \epsilon\) and \(s'_i \in V_P^{\text{abort}}\).

- If \(s_i\) is the last step of a variation or an extension and it is not an abort or goto step, then let \(s_j\) be the step that \(s_i\) extends or variates and \((s_j, \alpha_j, s'_j) \in \tau_P\) be the corresponding transition, it holds that \(s'_i = s'_j\).

- If \(s_i\) is a goto step and \(s_j \in U\) is the target step and let \((s_j, \alpha_j, s'_j)\) be the transition of \(P\) corresponding to the step \(s_j\), it holds that \(s'_i = s'_j\).

The annotation function \(\Lambda\) is defined as: if \(s_i\) is annotated by a set of annotations \(N\), then \(\Lambda(t_i) = N\).

**Example 1.** Figure 5 shows three textual use-cases \(U_1, U_2,\) and \(U_3\) and the corresponding UCAIs.

### 5.2 Resolution of the include relationship

We define the operation of resolution of includes – a transformation of UCAI to RUCA. This operation replaces include transitions with transitions of the included automata.
Definition 4 (Resolved use-case automaton–RUCA). Resolved use-case automaton (RUCA) is UCAI that does not contain any include action.

Definition 5 (Resolution of includes). Let $P$ be UCAI. Let $I$ be the set of use-case automata included in automaton $P$. The operation of resolution of includes ($\text{Res}$) transforms $P = (V_P, V_P^{\text{init}}, V_P^{\text{abort}}, V_P^{\text{succ}}, A_P, \tau_P)$ to RUCA $Q = (V_Q, V_Q^{\text{init}}, V_Q^{\text{abort}}, V_Q^{\text{succ}}, A_Q, \tau_Q)$ in the following way:

\[
\begin{align*}
V_Q &= V_P \cup \bigcup_{I \in I} V_{\text{Res}(U)} \\
V_Q^{\text{init}} &= V_P^{\text{init}} \\
V_Q^{\text{abort}} &= V_P^{\text{abort}} \\
V_Q^{\text{succ}} &= V_P^{\text{succ}} \\
A_Q &= A_P \setminus A_P^{\text{include}} \cup \bigcup_{U \in I} A_{\text{Res}(U)} \\
\tau_Q &= \tau_P \cup \tau_A \setminus \{\tau_I\}, \tau_I = (s, \text{inc}, s') \in \tau_P, \text{inc} \in A_P^{\text{inc}}, s, s' \in V_P
\end{align*}
\]

$\tau_A$ is defined as follows. Let $t_i = (s_i, \text{inc}, s_i') \in \tau_P, s_i, s_i' \in V_P$ be a transition of the automaton $P$ that contains an include action, let $Q_{\text{inc}}$ be UCAI associated with the include action $\text{inc}$ and $R = \text{Res}(Q_{\text{inc}})$ be the corresponding resolved use-case automaton. For every such a transition $t_i$, $\tau_A$ contains:

\[
\begin{align*}
(s_i, \epsilon, s_0) &\in V_R^{\text{init}} \\
(s_{\text{final}}, \epsilon, s_i') &\in V_R^{\text{succ}} \cup V_R^{\text{abort}}
\end{align*}
\]

Example 2. Figure 6a shows an example of UCAI $U_2$ from Figure 5 after the operation of resolution of includes.

5.3 Resolution of annotations

In textual use-cases, additional behavioral restrictions and consistence constraints are captured using annotations. Additional behavioral restrictions are captured using trace-on annotation pair and additional consistency properties are captured by create-use and open-close annotation pairs. We describe how these annotations define valuation of variables in transitions of the automaton, guard functions, and LTL formulae. Guard functions restrict sequences of transitions that the automaton captures and LTL formulae describe consistency requirements on the automaton.

Definition 6 (Valuation of states of RUCA). Let $P$ be RUCA and $X$ a set of variables. Valuation of transitions of $P$ over the set of variables $X$ is a function $\text{Val}_P : \tau_P \mapsto 2^X$ that maps each transition of $P$ to a set of variables. We denote each variable $v \in \text{Val}_P(s)$ as satisfied in a transition $s \in V_P$.

The set of variables $X_P$ is called variables of $P$ if $\forall x \in X_P: \exists v \in V_P$ such that $x \in \text{Val}_P(v)$. By $X_P = X_P \setminus \text{Val}(s)$ we denote the set of variables that are not satisfied in the transition $s$.

Definition 7 (Guard functions). Let $P$ be RUCA and $X_P$ a set of variables of $P$. Guard functions $\text{Guard}^+ : \tau_P \mapsto (2^{X_P})$ and $\text{Guard}^- : \tau_P \mapsto (2^{X_P})$ map each transition of $P$ to a set of variables.
The concept of guard functions is important for defining enabled transitions (Definition 13); how a guard function is constructed expresses the Definition 8.

**Definition 8 (Correspondence of annotations to valuation of a use-case automaton).** Let $P$ be a RUCA, $Af$ be an annotation function, and $N$ a set of all annotations of all transitions of $P$. Let $Val$ be a valuation function, and $Guard^+$ and $Guard^-$ guard functions. We say that $Val$, $Guard^+$, and $Guard^-$ correspond to $P$ and $Af$ when for each annotation $an \in N$ it holds:

- If $an$ is an annotation of the form $\text{trace} : id$, there is a variable $c_{id}$ such that $c_{id} \in Val(t)$.
- If $an$ is an annotation of the form $\text{on} : id$, there is a variable $c_{id}$ in $Guard^+(an)$ and for all transitions $t_u = (s_i, a_k, s_n)$, $s_n \neq s_j$, it holds that $c_{id} \in Guard^-(t_u)$.
- If $an$ is an annotation of the form $\text{create} : id$, there is a variable $c_{id}$ such that $c_{id} \in Val(t)$.
- If $an$ is an annotation of the form $\text{use} : id$, there is a variable $u_{id}$ such that $u_{id} \in Val(t)$.
- If $an$ is an annotation of the form $\text{open} : id$, there is a variable $o_{id}$ such that $o_{id} \in Val(t)$.
- If $an$ is an annotation of the form $\text{close} : id$, there is a variable $c_{id}$ such that $c_{id} \in Val(t)$.

Consequently, for $t_i \in \tau_p$ is $Guard^+(t_i) \cap Guard^-(t_i) = \emptyset$.

**Example 3.** Figure 6b shows RUCA with annotated transitions and Figure 6c shows this RUCA with valuations of transitions and guards. The transition $1a$ is annotated by a set of annotations $\{\text{on} : \text{abort1}\}$ and the other transition $i1$ from the input state of the transition $1a$ has no $\text{on} : id$ annotation. Hence, values of guard functions on these transitions are defined as follows: $Guard^+(1a) = \{c_{\text{abort1}}\}$, $Guard^-(1a) = \{\}$, $Guard^+(i1) = \{\}$, and $Guard^-(i1) = \{c_{\text{abort1}}\}$.

**Definition 9 (Consistency properties).** Let $P$ be RUCA, $Af$ be an annotation function, and $N$ be a set of all annotations of all states of $P$. A set of consistency properties associated with the automaton $P$ and annotation function $Af$ $LTL_P$ is a set of $LTL$ formulae defined as follows. For each annotation $an \in N$ $LTL_P$ contains:

- If $an$ is an annotation of form $\text{use} : id$, then $\neg u_{id} U c_{id} \in LTL_P$.
- If $an$ is an annotation of form $\text{create} : id$, then $\neg G(c_{id} \Rightarrow F(\neg u_{id})) \in LTL_P$.
- If $an$ is an annotation of form $\text{open} : id$, then $\neg o_{id} \Rightarrow F(c_{id} \in LTL_P$ and $o_{id} \Rightarrow \neg o_{id} U c_{id} \in LTL_P$.
- If $an$ is an annotation of form $\text{close} : id$, then $\neg c_{id} U o_{id} \in LTL_P$.

Note that these formulae represent consistency properties described in Section 4.1.

**Example 4.** For RUCA $P$ in Figure 6c we define the following $LTL$ formulae: $LTL_P = \{F(c_{\text{zoom}} \& \neg G u_{\text{zoom}}), \neg u_{\text{city}} U c_{\text{city}}, \neg u_{\text{zoom}} U c_{\text{zoom}}\}$.
5.4 Resolution of precedence relationship

Now, we define how automata capturing behavior of individual use-cases are serialized according to the precedence relationship.

**Definition 10 (Concatenated RUCA).** Let \( s = (R_1, R_2, \ldots, R_k) \) be an sequence of RUCA. Concatenated RUCA \( Q \) corresponding to \( s \) is defined as follows:

\[
\begin{align*}
V_Q &= \bigcup_{R_i} V_R \\
V_{\text{init}} &= V_{\text{init}}^R \\
V_{\text{abort}} &= V_{\text{abort}}^R \\
V_{\text{succ}} &= V_{\text{succ}}^R \\
A_Q &= \bigcup_{R_i} A_R \\
\tau_Q &= \bigcup_{R_i} \tau_R \cup \tau_A
\end{align*}
\]

\( \tau_A \) is defined as follows. Let \( (R_i, R_{i+1}) \) be a pair of subsequent resolved use-case automata in the sequence \( s \). Let \( \text{init}_{i+1} \) be the initial state of the automaton \( R_{i+1} \). For every such a pair and every final state \( \text{final}_i \in V_{\text{succ}}^R \cup V_{\text{abort}}^R \) of the automaton \( R_i \), there are transitions \((\text{succ}_i, \varepsilon, \text{init}_{i+1})\) and \((\text{final}_i, \varepsilon, \text{final}_i)\) \( \in \tau_A \).

Obviously, this definition stems from classical automata concatenation; the key enhancement here is the introduction of the transitions of the form \((\text{final}_i, \varepsilon, \text{final}_i)\), which corresponds to the semantics of \( \text{Prec} \). That is, \( U_i \) must occur before \( U_{i+1} \), hence all traces that reach \( U_{i+1} \) must go through \( U_i \). However, it is not required that \( U_{i+1} \) is executed after \( U_i \). There exist infinite traces that go through \( U_i \) and loop using the transition \((\text{final}_i, \varepsilon, \text{final}_i)\) thus never reaching \( U_{i+1} \).

**Example 5.** Figure 6d shows an example of concatenation of RUCA \( U_1 \) from Figure 5 and \( \text{Res}(U_2) \) from Figure 6a. The initial state of the resulting automaton is the initial state of \( U_1 \), abort and succeeded states of the resulting automaton are the same as abort and succeeded states of the automaton \( \text{Res}(U_2) \). The two automata are connected using transitions \( \rho_1 \) and \( \rho_2 \) going from the final states of the automaton \( U_1 \) to the initial state of the automaton \( \text{Res}(U_2) \). Then, there are looping transitions \( s_1 \) and \( s_2 \) going from each final state of \( U_1 \) back to this state. All these transitions contain the \( \varepsilon \) action.

**Definition 11 (Precedence Relation).** Precedence relation defined on a set of RUCA \( U \) \( \text{Prec} : U \times U \) is an antisymmetric and irreflexive relation, whose transitive closure \( \text{Prec}^* \) is antisymmetric and irreflexive as well. We say that \( U_i^R \) precedes \( U_j^R \) if \((U_i^R, U_j^R) \in \text{Prec} \). We say that \( U_k^R \) must be executed before \( U_l^R \) if \((U_k^R, U_l^R) \in \text{Prec}^* \).

**Definition 12 (Overall-behavior–OB).** Let \( U \) be a set of RUCA, let \( \text{Prec} \) be a precedence relation, and set \( S \) be the set of all permutations of RUCA from \( U \) ordered according to \( \text{Prec} \). The overall-behavior OB set with respect to \( U \) and \( \text{Prec} \) is the set of concatenated RUCA corresponding to members of \( S \).

**Example 6.** There are two permutations of use-cases in use-case specification in Figure 5 ordered according to specified precedences. That is, \((U_1, U_2, U_3)\) and \((U_1, U_3, U_2)\). Hence, the set OB for this specification consists of two automata.
5.5 Verification algorithm

In this section, we define the verification algorithm and related concepts.

**Definition 13 (Enabled transition).** Let $P$ be RUCA. Let $tr = v_0, a_0, v_1, a_1, ..., v_n$ be an alternating sequence of states and actions such that $t_i = (v_i, a_i, v_{i+1}) \in \tau_P$. The transition $t_i$ is enabled on $tr$ if all the transitions $t_j, j < i$ are enabled, for all $v^+ \in \text{Guard}^+(t_i)$ there exists $t_k, k \leq i$ such that $v^+ \in \text{Val}(t_k)$, and there is no $t_l, l \leq i$ such that for some $v^- \in \text{Guard}^-(t_i)$ it holds $v^- \in \text{Val}(t_l)$. If the transition is not enabled on $tr$, we say that it is disabled on $tr$.

**Example 7.** Consider the use-case automaton in Figure 6d and the sequence of transitions $(p1, i1, 1', 2a', 2a1', f2, 2b, 1b1)$. For the transition $p1$ both guard functions return the empty set and this transition is trivially enabled on $sq_1$. Next, $\text{Guard}^+(i1) = \{\}$ and $\text{Guard}^-(i1) = \{c_{\text{abort1}}\}$ and there is no predecessor $t_j$ of a transition $i1$ in the sequence $sq_1$ such that $c_{\text{abort1}} \in \text{Val}(t_j)$. Hence, the transition $i1$ is enabled on $sq_1$. Values of guard functions on transition $1'$ are the same and therefore this transition is also enabled on $sq_1$. Transitions $2a'$, $2a1'$, and $f2$ are trivially enabled on $sq_1$. Transition $1b$ is enabled on $sq_1$ because $\text{Guard}^+(1b) = \{c_{\text{abort2}}\}$ and $\text{Guard}^-(1b) = \{\}$. Hence, the transition $2a1'$ holds $\text{Val}(2a1') = \{c_{\text{abort2}}\}$.

Now, consider a sequence of transitions $(p1, 1a, 1a1)$. Similar to the previous example, the transition $p1$ is trivially enabled on $sq_2$. $\text{Guard}^+(1a) = \{c_{\text{abort1}}\}$ and there is no predecessor $s_j$ of the transition $1a$ in the sequence $sq_2$ for that $c_{\text{abort1}} \in \text{Val}(s_j)$. Hence, a transition $1a$ is disabled on $sq_2$. Both guard functions for a transition $1a1$ return the empty set, however, because a transition $1a$, which precedes the transition $1a1$, is disabled on $sq_2$; the transition $1a1$ is also disabled on $sq_2$.

**Definition 14 (Execution fragment).** An execution fragment of RUCA $P$ is an alternating sequence of states and actions $v_0, a_0, v_1, a_1, ..., v_n$ such that all transitions in the sequence $t_i = (v_i, a_i, v_{i+1}) \in \tau_P$ are enabled on $P$.

**Definition 15 (Execution trace).** An execution trace of RUCA $P$ is an execution fragment of use-case automaton $P$ that starts in the initial state of $P$ and is infinite or end in some final state $v_{\text{final}} \in V^\text{suc}\cup V^\text{abort}$ of the automaton $P$.

**Definition 16 (Consistent use-case).** A resolved use-case automaton $P$ is consistent if for all execution traces of the automaton $P$ all formulae from $LTL_P$ are satisfied.

**Verification algorithm.** The verification algorithm takes a set $U$ of use-cases (already parsed textual use-cases encoded in an internal form), and a precedence relation $Prec$ describing the precedence relationship among use-cases in $U$ as input. First, a static check of precedences and includes is done. If a cyclic dependency is found, the algorithm stops and returns not consistent.

Second, model-checking of temporal properties (using UCMC algorithm) is performed: UCIAI is built for each use-case in $U$ (Definition 3); the set of RUCAs is created by resolving all UCAs (Definition 5), then valuation of variables, guard functions (Definition 8), and consistency properties (Definition 9) are generated from annotations of RUCAs, the set OB is built (Definition 12) and then each automaton in OB is
model checked for consistency with generated LTL formulae (Definition 16). If all such automata are consistent, the algorithm returns consistent. If there is an concatenated RUCA in OB that is not consistent, there is an execution trace for which the LTL formula corresponding to a consistency property of the concatenated RUCA does not hold. In this case, the algorithm returns not consistent and provides further details which include (1) the steps of use-cases from \( U \) that correspond to this execution trace, and (2) the ordering of use-cases in OB.

The set of enabled transitions in a state of RUCA depends on states that precedes this transition in an execution trace, so that the classical LTL model-checking algorithm for Kripke structure based on Buchi automata cannot be applied in case of RUCA. Therefore, the state space is generated and the satisfaction LTL formulae is checked on-the-fly with respect to validity of guards. Using this approach, it is possible to check arbitrary LTL formulae. In the future work, we consider to let a user to specify arbitrary annotations that would only affect the valuation of variables and then let him/her to define arbitrary LTL formulae over these annotations. These LTL formulae would be translated into LTL formulae over variables introduced by the annotations and checked the same way as the automatically generated LTL formulae.

**Theorem 1 (Correctness of the verification algorithm).** Let \( U \) be the set of textual use-cases, \( G_{\text{prec}} \) be the graph describing a precedence relationship, and \( G_{\text{incl}} \) be the graph describing an include relationship. Assume that \( G_{\text{prec}} \) and \( G_{\text{incl}} \) do not contain cycles. Then, the algorithm returns consistent iff the specification consisting of a set of textual use-cases \( U \) and precedences among these use-cases does not contain any incorrectly used create, use, open or close annotation.

**Proof.** Correctness of the algorithm directly follows from the definitions above, semantics of textual use-cases, and semantics of the annotations. It can be proven in two steps. These are: (1) a proof that traces\(^6\) of transitions captured by the OB automata exactly correspond to the sequences of steps captured in the use-case specification when the annotations are not considered, and (2) a proof that execution traces of the OB automata correspond to the sequences of steps captured by the specification when the trace-on annotations are considered.

Then, from the Definition 9, it follows that LTL formulae generated from the create-use and open-close annotations correspond to semantics of these annotations. From this fact and the step (2), it follows that sequences of steps captured by the specification with correctly used annotations exactly correspond to the execution traces where all the generated LTL formulae are satisfied. Since there are no cyclic include dependencies and the number of variables is finite, the number of traces to explore is also finite and the algorithm eventually terminates. And thus the algorithm is correct.

Let us prove now the step (1). From the Definition 3 and the Definition 5 it follows that there is a sequence of steps that a given textual use-case describes iff there is a trace of transitions in RUCA corresponding to the use-case. From the Definition 10 it follows

\(^6\) The term trace in this context is defined in the same way as the execution trace (see Definition 15) with the modification that all the possible transitions are considered (not just the enabled ones).
that semantics of concatenation of RUCA corresponds to semantics of a precedence relation between textual use-cases. From the Definition 12 it follows that for each possible order of executions of the use-cases permissible by the specification there is a use-case automaton in the set of the OB automata that consists of the automata concatenated in this order.

Finally, let us prove the step (2). From the Definition 3 it follows that the annotations of steps of a use-case correspond to the annotations of traces of the RUCA. The trace-on annotations restrict sequences of steps captured by the specification. From this fact and (1), it follows that for each sequence of steps captured by the specification when the trace-on annotations are considered, there is a trace in a use-case automaton in the set of the OB automata. The trace is the execution trace iff for each transition annotated with the on:id annotation there is a transition annotated with the trace:id annotation before this transition. That is, there is no execution fragment not corresponding to a sequence of steps captured by the specification.

6 Summary and Future Work

We have developed means for verifying consistency of textual use-cases useful especially when use-cases are written iteratively by multiple authors. By introducing annotations to use-case steps, we can capture temporal dependencies among use-cases which is a foundation for further verification of temporal properties (based on LTL). As a key contribution, we have defined a formal behavior model (based on LTS) and defined its correspondence to textual use-case specification. A formal behavior model satisfying LTL formulae inferred from user annotations corresponds to a consistent use-case specification. Even though we have considered just two annotation pairs, the create-use and open-close pairs, our approach is applicable for other annotations as well, the semantics of which can be described by LTL. This is because we internally use LTL formulae to capture desired temporal properties. It should be noted that most of the examples in the text were taken from case studies of real-life use-cases [5].

Currently, we continue the development of REPROTOOL which integrates the verification method with analysis of natural language. As a future work we plan to tackle the following challenges:

- We plan to extend the palette of annotations in future and potentially to let users define their own annotations using arbitrary LTL-formulae.
- We could also implement asynchronous events in use-case specification. As pointed out by Larman [9] these events can be attached to multiple steps, e.g. “at any time” or “within a range of steps”.
- Our method would work even if we did not use any tools for processing natural language. Users could manually mark sentences as goto-, abort- or include-actions. However, due to the restrictions of the natural language in use-case specifications [3,9,20], we can benefit from NLP tools and thus automate this process. It should be also possible to infer the use-case step annotations from the text automatically. We intend to improve the currently employed NLP tools in REPROTOOL.
- The LTS constructed from use-cases can be used as an input for model-based testing and test-based modeling tools. This way, we intend to compare LTS (or similar
behavior model) extracted from a legacy application using test-based modeling approach [18], with LTS constructed using our method from use-case specification.

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References

Connectors as Designs: the Time Dimension

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Abstract. In this paper, we extend the design model for the coordination language Reo by introducing designs for timed connectors. Design is a key concept in Unifying Theories of Programming (UTP), which is used to describe the contract between programmer and client. The model developed in this paper specifies properties of timed component connectors properly. Implementation of the design model developed in \textsc{JTom} is provided.

Keywords: Coordination, Timed Reo Connector, UTP, Design, \textsc{JTom}

1 Introduction

Complex distributed applications are typically heterogeneous and geographically distributed, usually exploit communication infrastructures whose topology varies and components can, at any moment, connect to or detach from. The development of such distributed applications requires a coordination model that formalizes the orchestration among the components. Compositional coordination models and languages provide a formalization of the “glue code” that interconnects the constituent components and organizes the mutual interactions among them in a distributed processing environment. They support large-scale distributed applications by allowing construction of complex component connectors out of simpler ones. As an example, Reo \cite{4,8} offers a powerful glue language for implementation of coordinating component connectors. Primitive connectors called channels in Reo, such as synchronous channels and FIFO channels, can be composed to build circuit-like connectors which serve as the glue code to exogenously coordinate the behavior of components in distributed applications.

The problem that we identify and address in this paper is the time aspects of timed Reo connectors under the UTP semantic framework. UTP (Unifying Theories of Programming) is proposed by Hoare and He in \cite{13}, which can present a formal semantics for various programming languages and specification languages like Circus and timed Circus \cite{17,19}, TCOZ \cite{18}, rCOS \cite{14} and CSP \cite{12}. In \cite{16} we have shown that untimed Reo connectors can be translated into an essentially equivalent UTP design, i.e., a pair of predicates \( P \vdash Q \) where \( P \) specifies what the designer can rely on when the communicating operation is initiated by input to the connector, and \( Q \) is the condition on output that must be true when the communicating operation terminates. Connectors also admit intermediate observations on suitable occasions between initiation and termination. Additional variables and naming conventions are used to denote the results of such observations. In this paper the design model for untimed Reo is extended such
that the UTP specification of channels (and connectors) can involve timing constraints and we also show how the UTP design model for timed connectors are implemented in JTom (Tom\textsuperscript{3} built on top of Java)[2].

To give a semantic model for timed Reo, which clearly separates data and time, we need to choose an appropriate model of time. There are two typical models: discrete time and dense time. The discrete time model has been adopted in [19,18] on the semantics for timed Circus and TCOZ respectively. The dense time model, which is adopted in the timed data stream semantic model for Reo [6], seems to be more appropriate since it is very expressive and closer to the nature of time in the real world. For example, for a FIFO\textsubscript{1} channel, if we have a sequence of two inputs, the time moment for the output should be between the two inputs. If we use a discrete time model like ℤ, and have the first input at time point 1, then the second input can only happen at a time point greater than 2, i.e., at least 3. But in general, such timing constraints are not explicit for the input providers. Therefore, we choose the dense time model in this paper.

The formal semantics for Reo makes it possible to specify and analyze the behavior of a connector precisely. There are different formal semantics for Reo in literature. For example, a coalgebraic semantics for Reo in terms of relations on infinite timed data streams has been developed by Arbab and Rutten [6], but the causality between input and output is not clear in this semantics. An operational semantics for Reo using constraint automata is provided by Baier et al. [8]. However, modeling unbounded primitives or even bounded primitives with unbounded data domains is impossible with finite constraint automata. Bounded large data domains cause an explosion in the constraint automata model which becomes problematic. A model for Reo connectors based on the idea of coloring a connector with possible data flows to resolve synchronization and exclusion constraints is presented by Clarke et al. [9]. Unlike the coalgebraic and operational semantics, data sensitive behavior, which is supported by filter channels in Reo, are not captured in the coloring approach. The real-time aspects of Reo has been investigated in [5], which uses timed constraint automata (TCA) as the operational model for Reo connectors and provides a variant of LTL to serve as a specification formalism for timed Reo connectors. In [15], the TCA model is translated into mCRL\textsubscript{2} for model checking timed Reo connectors. However, these works suffer from the same problem as in the coalgebraic approach that the casuality between input and output is not clear.

Comparing with previous work on (both untimed and timed) Reo semantics, the UTP approach provides a family of algebraic operators for different kinds of composition of connectors (merger, replicator and flow-through composition), which can be used to interpret the composition of connectors more explicitly than other approaches, such as TDS in the coalgebraic semantics. Moreover, by separation of the assumption (input) and the commitment (output) in the design model in our approach, the connector behavior becomes more clear and easy to be composed. The UTP approach also makes it possible to check connector properties by assume-guarantee reasoning. Properties of

\[ \text{Tom} \text{ is a powerful and efficient pattern matching engine on top of conventional programming languages like C or Java.} \]
a complex connector can be decomposed into properties of its subconnectors and each subconnector can be checked separately.

The paper is structured as follows. After this general introduction, we briefly summarize the coordination language Reo and introduce the notion of design in UTP being used throughout the rest of the paper in Section 2. Section 3 presents the model of observations on connector nodes. Section 4 briefly summarizes the UTP design model for basic (untimed and timed) Reo channels and timed Reo circuits. In Section 5, we discuss the implementation of the design model in Tom. Finally, Section 6 concludes with some further research directions.

2 Preliminaries

In this section, we briefly review basic concepts of Reo and UTP design. The discussion on Reo is mainly based on [4,8] and the overview of UTP design is from [13].

2.1 Reo

Reo [4] is a channel-based exogenous coordination language wherein complex coordinators, called connectors, are compositionally constructed from simpler ones. We summarize only the main concepts in Reo here. Further details about Reo can be found in [4,8].

A Reo connector consists of a network of primitive connectors, called channels. A connector provides the protocol that controls and organizes the communication, synchronization and cooperation among the components/services that communicate through the connector. Each channel has two channel ends: source and sink. A source channel end accepts data into its channel, and a sink channel end dispenses data out of its channel. It is possible for the ends of a channel to be both sinks or both sources. Each channel end can be connected to at most one component instance at any given time. Figure 1 shows the graphical representation of some simple channel types in Reo whose composition allows for expressing a rich set of coordination patterns [4].

A *FIFO1 channel* represents an asynchronous channel with one buffer cell which is empty if no data item is shown in the box (this is the case in Figure 1). If a data element $d$ is contained in the buffer of a FIFO1 channel then $d$ is shown inside the box in its graphical representation. A *synchronous channel* has a source and a sink end and no buffer. It accepts a data item through its source end iff it can simultaneously...
dispense it through its sink. A *lossy synchronous channel* is similar to a synchronous channel except that it always accepts all data items through its source end. The data item is transferred if it is possible for the data item to be dispensed through the sink end, otherwise the data item is lost. *Synchronous drain* has two source ends and no sink end. A synchronous drain can accept a data item through one of its ends iff a data item is also available for it to simultaneously accept through its other end as well, and all data accepted by the channel are lost. More exotic channels permitted in Reo will be omitted here and can be found in [4]. Note that the set of channel types is not fixed in Reo, and new ones can be defined freely by users with their own interaction policies.

Complex connectors are constructed by composing simpler ones via the *join* and *hiding* operations. Channels are joined together in nodes. A node consists of a set of channel ends. The set of channel ends coincident on a node $A$ is disjointly partitioned into the sets of source and sink channel ends that coincide on $A$. Nodes are categorized into *source*, *sink* and *mixed nodes*, depending on whether all channel ends that coincide on a node are source ends, sink ends or a combination of the two. The hiding operation is used to hide the internal topology of a component connector. The hidden nodes can no longer be accessed or observed from outside. A complex connector has a graphical representation, called a *Reo circuit*, whose behavior is formalized by means of the data-flow at its sink and source nodes. Intuitively, the source nodes of a circuit are analogous to the input ports, and the sink nodes to the output ports of a component, while mixed nodes are its hidden internal details.

![Complex Connectors](image)

*Fig. 2. Source, Sink and Mixed Nodes in Reo*

A component can write data items to a source node that it is connected to. The write operation succeeds only if all (source) channel ends coincident on the node accept the data item, in which case the data item is transparently written to every source end coincident on the node. A source node, thus, acts as a replicator. A component can obtain data items, by an input operation, from a sink node that it is connected to. A take operation succeeds only if at least one of the (sink) channel ends coincident on the node offers a suitable data item; if more than one coincident channel end offers suitable data items, one is selected non-deterministically. A sink node, thus, acts as a non-deterministic merger. A mixed node non-deterministically selects and takes a suitable data item offered by one of its coincident sink channel ends and replicates it into all of its coincident source channel ends.

### 2.2 A Theory of Designs

A direct and obvious way to represent the observable behavior of a connector is to model it as a relation on its inputs and outputs. Because the inputs / outputs take place...
through the nodes of the connector, sequences of data items that pass through a node
together with the moments in time that the data items are observed emerge as the key
building blocks for describing connector behavior.

In our semantic model, we use two auxiliary variables \( ok \) and \( ok' \) to analyze
explicitly the phenomena of communication initialization and termination. The variable
\( ok \) stands for a successful initialization and the start of a communication. When \( ok \)
is \textit{false}, no observation can be made. The variable \( ok' \) denotes the observation that
the communication has either terminated or reached an intermediate stable state. The
communication is divergent when \( ok' \) is \textit{false}. The observational semantics for a Reo
connector is described by a design, i.e., a relation expressed as \( P \vdash Q \), where \( P \) is the
predicate specifying the relationship among the observations on the source nodes of the
connector, and \( Q \) is the predicate specifying the condition that should be satisfied by
the observations on the sink nodes of the connector. Such a design \( P \vdash Q \) is defined as
follows:

**Definition 1.** A design is a pair of predicates \( P \vdash Q \), where neither predicate contains
\( ok \) or \( ok' \), and \( P \) has only input variables. It has the following meaning:

\[
P \vdash Q =_{df} (ok \land P \Rightarrow ok' \land Q)
\]

The separation of condition on inputs from condition on outputs in our model allows
us to write a specification that has a more generous precondition than simply the domain
of a relation used as a specification. We are allowed to assume that the condition on
inputs holds, and we have to satisfy the condition on its outputs. Moreover, we can
can rely on the behavior of a connector having been started, but we must ensure that it
terminates. If the condition on inputs does not hold, or the connector does not start its
behavior, we are not committed to establish the condition on the outputs nor even to
make the connector behavior terminate.

A facility to select between alternatives according to the truth or falsehood of some
guard \( b \) is necessary for non-trivial behavior. Conditional expression \( P \prec b \Succ Q \) describes
a system that behaves like \( P \) if the initial value of \( b \) is \textit{true}, or like \( Q \) otherwise. It can
be defined as follows:

**Definition 2.** The conditional expression is defined as follows:

\[
P \prec b \Succ Q =_{df} \text{true} \vdash (b \land P \lor \neg b \land Q)
\]

The sequential composition \( P ; Q \) denotes a system that first executes \( P \), and when
\( P \) terminates executes \( Q \). This system is defined via existential quantification to hide
its intermediate observation, and to remove the variables that record this observation
from the list of free variables of the predicate. To accomplish this hiding, a fresh set
of variables \( v_0 \) is used to denote the intermediate observation. These fresh variables
replace the input variables \( v \) of \( Q \) and the output variables \( v' \) of \( P \), thus the output
alphabet of \( P (\text{out}_a P) \) and the input alphabet of \( Q (\text{in}_a Q) \) must be the same.

**Definition 3.** Let \( \text{out}_a P = \{v'\} \), \( \text{in}_a Q = \{v\} \), then

\[
P (\text{in} : u ; \text{out} : v') ; Q (\text{in} : v ; \text{out} : w) =_{df} \exists v_0 \text{\Leftrightarrow} P (\text{in} : u ; \text{out} : v_0) \land Q (\text{in} : v_0 ; \text{out} : w)
\]
If the conditional and sequential operators are applied to designs, the result is also a design. This follows from the laws below.

\[ (P_1 \vdash Q_1) \triangleleft b \triangleright (P_2 \vdash Q_2) = ((P_1 \triangleleft b \triangleright P_2) \vdash (Q_1 \triangleleft b \triangleright Q_2)) \]
\[ (P_1 \vdash Q_1); (P_2 \vdash Q_2) = (P_1 \land \neg(Q_1; \neg P_2) \vdash (Q_1; Q_2)) \]

In UTP, we have the well-known property for refinement, which is established by the following definition:

**Definition 4.** \([ (P_1 \vdash Q_1) \subseteq (P_2 \vdash Q_2) ] \) **iff** \([ P_1 \Rightarrow P_2 ] \land [ P_1 \land Q_2 \Rightarrow Q_1 ] \)

### 3 Observations on Connectors

To specify inputs and outputs on connectors explicitly, for a connector \( R \), we use \( \text{in}_R \) and \( \text{out}_R \) to denote the observations on its source nodes and sink nodes, respectively, instead of using unprimed variables for initial observations (inputs) and primed variables for subsequent ones (outputs) as in [13] (the definition of \( \text{in}_R \) and \( \text{out}_R \) will be given later).

For every node \( N \) in a connector \( R \), the corresponding observation on \( N \) is given by a *timed data sequence*, which is defined as follows:

Let \( D \) be an arbitrary set, the elements of which are called data elements. The set \( DS \) of data sequences is defined as \( DS = D^* \), i.e., the set of all sequences \( \alpha = (\alpha(0), \alpha(1), \alpha(2), \cdots) \) over \( D \).

Let \( \mathbb{R}_+ \) be the set of non-negative real numbers, which in the present context can be used to represent time moments. For a sequence \( s \), the length of \( s \) is denoted by \(|s|\). If \( s \) is an infinite sequence, then \(|s| = \infty \). Let \( \mathbb{R}_+^* \) be the set of sequences \( a = (a(0), a(1), a(2), \cdots) \) over \( \mathbb{R}_+ \), and for all \( a, b \) in \( \mathbb{R}_+^* \), if \(|a| = |b|\), then

\[
a < b \quad \text{iff} \quad \forall 0 \leq n < |a|, a(n) < b(n) \\
a \leq b \quad \text{iff} \quad \forall 0 \leq n < |a|, a(n) \leq b(n)
\]

For a sequence \( a = (a(0), a(1), a(2), \cdots) \in \mathbb{R}_+^* \), and \( t \in \mathbb{R}_+ \), \( a[+t] \) is a sequence defined as follows:

\[
a[+t] = (a(0) + t, a(1) + t, a(2) + t, \cdots)
\]

Furthermore, the element \( a(n) \) in a sequence \( a = (a(0), a(1), a(2), \cdots) \) can also be expressed in terms of derivatives \( a(n) = a^{(n)}(0) \), where \( a^{(n)} \) is defined by

\[
a^{(0)} = a, \quad a^{(1)} = (a(1), a(2), \cdots), \quad a^{(k+1)} = (a^{(k)}(k+1))
\]

The set \( TS \) of time sequences is defined as

\[
TS = \{a \in \mathbb{R}_+^* \mid (\forall 0 \leq n < |a|, a(n) < a(n + 1)) \land \ (|a| = \infty \Rightarrow \forall t \in \mathbb{R}_+, \exists k \in \mathbb{N}. a(k) > t)\}
\]

Thus, a time sequence \( a \in TS \) consists of increasing and diverging time moments \( a(0) < a(1) < a(2) < \cdots \).
For a sequence $a$, the two operators $a^R$ and $\overrightarrow{a}$ are used to denote the reverse and tail of $a$ respectively, which are defined as:

$$a^R = \begin{cases} () & \text{if } a = () \\ (a')^R(a(0)) & \text{if } a = (a(0))^-a' \end{cases}$$

$$\overrightarrow{a} = \begin{cases} () & \text{if } a = () \\ a' & \text{if } a = (a(0))^-a' \end{cases}$$

where $^-\!$ is the concatenation operator on sequences. The concatenation of two sequences produces a new sequence that starts with the first sequence followed by the second sequence.

The set $TDS$ of timed data sequences is defined as $TDS \subseteq DS \times TS$ of pairs $\langle \alpha, a \rangle$ consisting of a data sequence $\alpha$ and a time sequence $a$ with $|\alpha| = |a|$. Similar to the discussion in [6], timed data sequences can be alternatively and equivalently defined as (a subset of) $(D \times \mathbb{R}^+)^*$ because of the existence of the isomorphism

$$\langle \alpha, a \rangle \mapsto (\langle \alpha(0), a(0) \rangle, \langle \alpha(1), a(1) \rangle, \langle \alpha(2), a(2) \rangle, \cdots)$$

The occurrence of a data item $d$ at some node $N$ of a connector (i.e., taking $d$ from $N$ or writing $d$ to $N$) is modeled by an element in the timed data sequence for that node, i.e., a pair of a data element and a time moment.

4 Reo and its Design Model

In this section we provide an overview on how Reo connectors can be modeled as UTP designs. We first see how primitive untimed channels in Reo are specified as designs, and then study the design model of timed channels. Finally we show how composite connectors can be constructed from simpler ones structurally.

We use $\mathcal{WD}$ for a predicate of well-defined $TDS$ types. In other words, we define the behavior only for valid sequences expressed via a predicate $\mathcal{WD}$. Then, every connector $R$ can be represented by the design $P(in_R) \vdash Q(in_R, out_R)$ as follows:

$$\text{con: } R(in : in_R; out : out_R)$$

$$\text{in: } P(in_R)$$

$$\text{out: } Q(in_R, out_R)$$

where $R$ is the name of the connector, $P(in_R)$ is the condition that should be satisfied by inputs $in_R$ on the source nodes of $R$, and $Q(in_R, out_R)$ is the condition that should be satisfied by outputs $out_R$ on the sink nodes of $R$. Let $N_{in}$ and $N_{out}$ be the set of source and sink node names of $R$, respectively, then $in_R$ and $out_R$ are defined as the following mappings from the corresponding sets to $TDS$:

$$in_R : N_{in} \to TDS$$

$$out_R : N_{out} \to TDS$$
4.1 Designs for Primitive Reo Channels

We now start by presenting a few examples of basic channels in Reo and their design model. Discussions on more channel types can be found in [16].

**FIFO channels.** The simplest form of an asynchronous channel is a FIFO channel with one buffer cell, which is denoted as \( \text{FIFO1} \). A \( \text{FIFO1} \) channel with source end \( A \) and sink end \( B \) is graphically represented by \( A \rightarrow \leftarrow B \). The design model for a \( \text{FIFO1} \) channel is given as follows:

\[
\begin{align*}
\text{con} & : \ \text{FIFO1}(in : (A \mapsto \langle \alpha, a \rangle); out : (B \mapsto \langle \beta, b \rangle)) \\
\text{in} & : \ \mathcal{WD}(\alpha, a) \\
\text{out} & : \ \mathcal{WD}(\beta, b) \land \beta = \alpha \land a < b \land (b^{R})^R < a^R
\end{align*}
\]

For a FIFO1 channel, when the buffer is not filled, the input is accepted without immediately outputting it. The accepted data item is kept in the internal FIFO buffer of the channel. The next input can happen only after an output occurs. Note that here we use \((b^{R})^R < a^R\) to represent the relationship between the time moments for outputs and their corresponding next inputs. This notation can be simplified to \(b < a\) if we consider infinite sequences of inputs and outputs.  

For the \( \text{FIFO1} \) channel \( A \rightarrow \leftarrow B \) where the buffer contains the data element \( e \), the communication can be initiated only if the data element \( e \) can be taken via the sink end. In this case, we denote the channel by \( \text{FIFO1}e \) and have its design model as follows:

\[
\begin{align*}
\text{con} & : \ \text{FIFO1}(in : (A \mapsto \langle \alpha, a \rangle); out : (B \mapsto \langle \beta, b \rangle)) \\
\text{in} & : \ \mathcal{WD}(\alpha, a) \\
\text{out} & : \ \mathcal{WD}(\beta, b) \land \beta = (e)^\sim \land a < (b^R)^R < a
\end{align*}
\]

**Synchronous channels.** A synchronous channel transfers the data without delay in time. So it behaves just as the identity function. The pair of I/O operations on its two ends can succeed only simultaneously. A synchronous channel with source end \( A \) and sink end \( B \) is graphically represented as \( A \rightarrow B \). The design model for a synchronous channel is as follows:

\[
\begin{align*}
\text{con} & : \ \text{Sync}(in : (A \mapsto \langle \alpha, a \rangle); out : (B \mapsto \langle \beta, b \rangle)) \\
\text{in} & : \ \mathcal{WD}(\alpha, a) \\
\text{out} & : \ \mathcal{WD}(\beta, b) \land \beta = \alpha \land b = a
\end{align*}
\]

A lossy synchronous channel (graphically depicted as \( A \rightarrow B \)) is similar to a normal synchronous channel, except that it always accepts all data items through its source end. If it is possible for it to simultaneously dispense the data item through its

---

4 Note that \((b^{R})^R\) denotes the sequence obtained by removing the \( n \)-th element from a sequence \( b \) with length \( n \). For infinite sequences, we have \((b^{R})^R = b\).
sink end, the channel transfers the data item; otherwise the data item is lost. Its design model is given as follows:

\[
\text{con} : \text{LossySync}(\text{in} : (A \mapsto \langle \alpha, a \rangle); \text{out} : (B \mapsto \langle \beta, b \rangle))
\]

\[
\text{in} : \ WD\langle \alpha, a \rangle
\]

\[
\text{out} : \ WD\langle \beta, b \rangle \land L(\langle \alpha, a \rangle, \langle \beta, b \rangle)
\]

where

\[
L(\langle \alpha, a \rangle, \langle \beta, b \rangle) \equiv (\beta = ( ) \land b = ( )) \lor \left( a(0) \leq b(0) \land \begin{cases} \alpha(0) = \beta(0) \land L(\langle \alpha, a \rangle, \langle \beta, b \rangle) & \text{if } a(0) = b(0) \\ L(\langle \alpha, a \rangle, \langle \beta, b \rangle) & \text{if } a(0) < b(0) \end{cases} \right)
\]

An exotic channel in Reo is the synchronous drain \( A \leftrightarrow B \) that has two source ends. Because a drain has no sink end, no data value can ever be obtained from this channel. Thus, all data accepted by this channel are lost. A synchronous drain accepts a data item through one of its ends iff a data item is also available for it to simultaneously accept through another end.

\[
\text{con} : \text{SyncDrain}(\text{in} : (A \mapsto \langle \alpha, a \rangle, B \mapsto \langle \beta, b \rangle); \text{out} : ( ))
\]

\[
\text{in} : \ WD\langle \alpha, a \rangle \land WD\langle \beta, b \rangle \land a = b
\]

\[
\text{out} : \text{true}
\]

A filter channel \( A \rightarrow \{p\} \rightarrow B \) specifies a filter pattern \( p \) which is a set of data values. It transfers only those data items that match with the pattern \( p \) and loses the rest. A write operation on the source end succeeds only if either the data item to be written does not match with the pattern \( p \) or the data item matches the pattern \( p \) and it can be taken synchronously via the sink end of the channel.

\[
\text{con} : \text{Filter} p(\text{in} : (A \mapsto \langle \alpha, a \rangle); \text{out} : (B \mapsto \langle \beta, b \rangle))
\]

\[
\text{in} : \ WD\langle \alpha, a \rangle
\]

\[
\text{out} : \ WD\langle \beta, b \rangle \land F(\langle \alpha, a \rangle, \langle \beta, b \rangle)
\]

where

\[
F(\langle \alpha, a \rangle, \langle \beta, b \rangle)
\]

\[
\equiv \begin{cases} \beta = ( ) \land b = ( ) & \text{if } \alpha = ( ) \land a = ( ) \\ \beta(0) = a(0) \land b(0) = a(0) \land F(\langle \alpha, a \rangle, \langle \beta, b \rangle) & \text{if } a(0) \in p \\ F(\langle \alpha, a \rangle, \langle \beta, b \rangle) & \text{if } a(0) \notin p \end{cases}
\]

4.2 Designs for Reo Timer Channels

We now describe the design models of a few timer channels in Reo that can serve to measure the time between two events and produce timeout signals.
The source end of a t-timer $A \xrightarrow{t} B$ channel accepts any input value $d$ and returns on its sink end $B$ a timeout signal after a delay of $t$ time units.

$$\text{con} : \quad \text{Timer}(\text{in} : (A \mapsto \langle \alpha, a \rangle); \text{out} : (B \mapsto \langle \beta, b \rangle))$$

$$\text{in} : \quad WD(\alpha, a) \land a[+]t \leq \overline{a}$$

$$\text{out} : \quad WD(\beta, b) \land b \in \{\text{timeout}\} \lor b = a[+]t$$

An exciting behavior for a similar timer channel $A \xrightarrow{t} B$ is to produce a timeout after a delay $t$ for all inputs, even if the inter-arrival time of the inputs is less than $t$ (maybe for up to $n$ turns on its sink end $B$ after a delay when a special “off” value is consumed through its source end. A $t$-timer with the $\text{off}$-option $A \xrightarrow{t} B$ allows the timer to be stopped before the expiration of its delay when a special “off” value is consumed through its source end.

$$\text{con} : \quad \text{OFFTimer}(\text{in} : (A \mapsto \langle \alpha, a \rangle); \text{out} : (B \mapsto \langle \beta, b \rangle))$$

$$\text{in} : \quad WD(\alpha, a) \land \forall i < |a|.(a[i]+t(i) \leq \overline{a}(i) \lor \overline{a}(i) = \text{off})$$

$$\text{out} : \quad WD(\beta, b) \land \beta \in \{\text{timeout}\} \lor \text{OFF}((\alpha, a), \langle \beta, b \rangle)$$

where

$$\text{OFF}((\alpha, a), \langle \beta, b \rangle)$$

$$\equiv \begin{cases} \langle \beta, b \rangle = \langle (), () \rangle & \text{if } (\alpha, a) = \langle (), () \rangle \\ \text{OFF}((\overline{\alpha}, \overline{a}), \langle \overline{\beta}, \overline{b} \rangle) \land b(0) = a(0) + t & \text{if } a(0) + t \leq a(1) \lor |a| = 1 \\ \text{OFF}((\alpha, a), \langle \beta, b \rangle) & \text{if } a(1) < a(0) + t \land a(1) = \text{off} \end{cases}$$

Similarly, the $\text{reset}$-option allows the timer to be reset to 0 after it has been activated when a special “reset” value is consumed through its source end. A $t$-timer with the $\text{reset}$-option is graphically represented as $A \xrightarrow{t} \overline{D} \xrightarrow{t} B$ and its design model is given as follows:

$$\text{con} : \quad \text{RESETTimer}(\text{in} : (A \mapsto \langle \alpha, a \rangle); \text{out} : (B \mapsto \langle \beta, b \rangle))$$

$$\text{in} : \quad WD(\alpha, a) \land \forall i < |a|.(a[i]+t(i) \leq \overline{a}(i) \lor \overline{a}(i) = \text{reset})$$

$$\text{out} : \quad WD(\beta, b) \land \beta \in \{\text{timeout}\} \lor \text{RT}((\alpha, a), \langle \beta, b \rangle)$$

where

$$\text{RT}((\alpha, a), \langle \beta, b \rangle)$$

$$\equiv \begin{cases} \langle \beta, b \rangle = \langle (), () \rangle & \text{if } (\alpha, a) = \langle (), () \rangle \\ \text{RT}((\overline{\alpha}, \overline{a}), \langle \overline{\beta}, \overline{b} \rangle) \land b(0) = a(0) + t & \text{if } a(0) + t \leq a(1) \lor |a| = 1 \\ \text{RT}((\overline{\alpha}, \overline{a}), \langle \beta, b \rangle) & \text{if } a(1) < a(0) + t \land a(1) = \text{reset} \end{cases}$$
A timer with early expiration makes the timer produce its timeout signal through its sink and reset itself when it consumes a special "expire" value through its source end. A $t$-timer with early expiration is graphically represented as $A \leftarrow (\{} \rightarrow B$ and its design model is given as follows:

$$\text{con} : \text{EXPIRETimer}(in : (A \rightarrow (\alpha, a)); out : (B \rightarrow (\beta, b)))$$

$$\text{in} : \quad \forall i < |a|, (a[i+t](i) \leq \overrightarrow{a}(i) \lor \exists i = \text{expire})$$

$$\text{out} : \quad \forall i < |b|, (b(i) \in \{\text{timeout}\}^* \land \exists t \in \text{ET}(\langle \alpha, a \rangle, \langle \beta, b \rangle))$$

where

$$\text{ET}(\langle \alpha, a \rangle, \langle \beta, b \rangle) \equiv \begin{cases} \langle \beta, b \rangle = \langle (), () \rangle & \text{if } \langle \alpha, a \rangle = \langle (), () \rangle \\ \text{ET}(\langle \overrightarrow{\alpha}, \overrightarrow{a} \rangle, \langle \overrightarrow{\beta}, \overrightarrow{b} \rangle) \land b(0) = a(0) + t & \text{if } a(0) + t \leq a(1) \lor |a| = 1 \\ \text{ET}(\langle \overrightarrow{\alpha}, \overrightarrow{a} \rangle, \langle \overrightarrow{\beta}, \overrightarrow{b} \rangle) \land b(0) = a(1) & \text{if } a(1) < a(0) + t \land \alpha(1) = \text{expire} \end{cases}$$

### 4.3 Designs for Reo Circuits

Complex connectors in Reo are organized as a network of channels, which are called Reo circuits. Since basic channels can be modeled by designs, their composition can be defined as design composition, and the resulting connector is still a design. According to the node types in Reo, we have three types of composition, which can be captured by the three configurations as shown in Figure 3.

![Fig. 3. Connector composition](image)

Different channel ends can be joined together into new nodes. If all channel ends coincident on a node are source (sink) channel ends, the node is called a source (sink) node. Figure 3(1) shows the case of flow-through composition of connectors. For the two connectors $R_1$ and $R_2$, suppose one sink node of $R_1$ and one source node of $R_2$ are joined together into a node $B$. In this case, the node $B$ is called a mixed node which behaves as a self-contained pumping station. When we compose connectors, we want the events on the mixed nodes to happen silently and automatically whenever they can, without the participation or even the knowledge of the environment of the connector. Such mixed nodes are hidden (encapsulated) by using existential quantification. Figure
3(2) shows the case of merging two sink nodes of the connectors $R_1$ and $R_2$. In this case the node $C$ behaves as a merger. An attempt to take a data item from the node succeeds when at least one of its coincident channel ends has a suitable data to offer, in which case the suitable data available through one of its coincident channel ends is non-deterministically selected to be taken. Figure 3(3) shows the case of merging two source nodes of the connectors $R_1$ and $R_2$. In this case the node acts as a replicator. Writing a data item to a source node succeeds when all of the coincident channel ends are capable of accepting the data item simultaneously, in which case the data item is atomically copied into every source ends coincident on $A$. Due to the length limitation, we will only give the definition for merging sink nodes operation here. Details for the other composition operators can be found in [16].

Let $(\gamma_i, c_i)$ for $i = 1, 2$ be the timed data sequences on the sink node $C$ in $R_1$ and $R_2$, respectively. Then by merging the two sink nodes, the resulting connector is denoted by $R = R_1 \times_C R_2$, and the corresponding design is given as

$$\begin{align*}
\text{con} & : R(\text{in} : \bigcup_{i=1,2} in_{R_i}, \text{out} : \bigcup_{i=1,2} (\text{out}_{R_i} \setminus \{ C \mapsto (\gamma_i, c_i) \}) \cup \{ C \mapsto (\gamma, c) \}) \\
\text{in} & : \bigwedge_{i=1,2} P_i(in_{R_i}) \\
\text{out} & : \text{WD}(\gamma, c) \land \\
& \exists (\gamma_1, c_1), (\gamma_2, c_2), (\bigwedge_{i=1,2} Q_i(in_{R_i}, \text{out}_{R_i})) \land M((\gamma_1, c_1), (\gamma_2, c_2), (\gamma, c))
\end{align*}$$

In this definition, the ternary relation $M$ is defined as

$$M((\gamma_1, c_1), (\gamma_2, c_2), (\gamma, c)) = \begin{cases} 
(\gamma, c) = (\gamma_1, c_1) & \text{if } |(\gamma_2, c_2)| = 0 \\
(\gamma, c) = (\gamma_2, c_2) & \text{if } |(\gamma_1, c_1)| = 0 \\
c_1(0) \neq c_2(0) \land \\
\gamma(0) = \gamma_1(0) \land c(0) = c_1(0) \land \\
M((\gamma_1^1, c_1^1), (\gamma_2, c_2), (\gamma_2, c_2)) & \text{if } c_1(0) < c_2(0) \\
\gamma(0) = \gamma_2(0) \land c(0) = c_2(0) \land \\
M((\gamma_1, c_1), (\gamma_2^2, c_2^2), (\gamma_2, c_2)) & \text{if } c_2(0) < c_1(0) \\
\text{otherwise}
\end{cases}$$

In this relation $M$, two timed data sequences $(\gamma_1, c_1)$ and $(\gamma_2, c_2)$ are merged together into one single timed data sequence $(\gamma, c)$ where the order of elements in the sequence is decided by the time moments. Furthermore, if we have three timed data sequences $(\gamma_1, c_1)$, $(\gamma_2, c_2)$ and $(\gamma, c)$, such that $M((\gamma_1, c_1), (\gamma_2, c_2), (\gamma, c))$ is satisfied, then we say that $(\gamma, c) = \text{merge}((\gamma_1, c_1), (\gamma_2, c_2))$. Moreover, the merging operation can also be defined on the data and time sequences separately: $\gamma = \text{merge}(\gamma_1, \gamma_2)$ and $c = \text{merge}(c_1, c_2)$.

**Example 1.** We consider the circuit given in Figure 4 as a simple example. This circuit ensures the lower bound “$> t$” for a take operation on node $B$; it yields a FIFO1
channel that guarantees every data item will remain in its buffer at least $t$ time units. The design model for this connector can be easily obtained from the composition of the basic channels:

$$
\text{con} : \quad \text{LowerboundedFIFO1}(in : (A \mapsto (\alpha, a)); out : (B \mapsto (\beta, b))) \\
\text{in} : \quad \text{WD}(\alpha, a) \land a[+t] \leq \bar{a} \\
\text{out} : \quad \text{WD}(\beta, b) \land \beta = \alpha \land a[+t] < b \land (b^R - \bar{a}) < \bar{a}
$$

Example 2. A useful connector that can be used to model a real-time network is a timed FIFO$n$ that delays every input for $t$ time units, even if the inter-arrival time of the inputs is less than $t$ (for up to $n$ such inputs). Such a connector can not be obtained by just composing $n$ timed FIFO1 channels. However, it is still easy to be constructed by using $\rightarrow \leftarrow$. Figure 5 shows an example of such a timed FIFO3 connector. We can parameterize this connector to have as many buffers as we want simply by inserting more (or fewer) FIFO1 channels between nodes $A$ and $F$, as required. The design model for this connector is as follows$^5$:

$$
\text{con} : \quad \text{timedFIFO}(in : (A \mapsto (\alpha, a)); out : (B \mapsto (\beta, b))) \\
\text{in} : \quad \text{WD}(\alpha, a) \\
\text{out} : \quad \text{WD}(\beta, b) \land \beta = \alpha \land b = a[+t] \land b < a(n)
$$

$^5$ To simplify the expression we consider infinite sequences of inputs and outputs here. For finite sequences with length $k$, the only difference is to replace the condition $b < a(n)$ by the relationship that $b(i) < a(n + i)$ for any $i \leq k - n$. 

Fig. 4. Lower Bounded FIFO1 Channel

Fig. 5. Timed FIFO3 Connector
Example 3. A variant of FIFO1 is called expiring FIFO1, denoted by $A - [\square] \tau B$, where a data item is lost if it is not taken out of the buffer through the sink end of the channel within $t$ time units after it enters through its source end. Figure 6 shows on the left how an expiring FIFO1 channel can be constructed out of a normal FIFO1 channel and a timer.

![Fig. 6. Expiring FIFO1 Channel](image)

By composing the channels together (and after some equational transformations to simplify the expression), we can get the design model for the expiring FIFO1 as:

\[
\text{con : ExpiringFIFO1(in : } (A \mapsto \{\alpha, a\}); \text{ out : } B \mapsto \{\beta, b\})
\]

\[
\text{in : } WD(\alpha, a) \land a[+t] \leq \bar{a}
\]

\[
\text{out : } \exists(\pi, h). WD(\beta, b) \land WD(\pi, h) \land
\]

\[
a \prec \text{merge}(b, h) \land (\text{merge}(b, h))^R \leq \bar{a} \land L((\alpha, \text{merge}(b, h)), (\beta, b)) \land
\]

\[
L((\text{timeout}^*, a[+t]), (\pi, h))
\]

5 Implementation

Higher-level languages like Maude [10] and Tom [11] offer suitable means to quickly prototype our design model for Reo connectors. In contrast to dedicated languages like Java, the advantages of such languages include easy prototyping, clear separation between syntax and semantics, and high-level mechanisms like strategies for meta-control of the executions of the programs. The reason why we choose Tom instead of Maude (the language of choice in [3] for implementation of untimed Reo connectors) for implementing the design model for timed connectors is that Tom is much more flexible and support pattern matching against native data-structures like objects or records. This is a strong point since we can take advantage of existing efficient implementations of data structures like ArrayList or HashMap and the corresponding functions to manipulate the basic ingredients in the theory of connectors, which are sequences and streams.

\*

Here we use $\text{timeout}^*$ to denote the sequence $s \in \{\text{timeout}\}^*$.

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In this section we focus on how the syntax and the semantics of timed connector are implemented in JTom. This is of concern especially for future developers and should be seen as a short documentation for a library providing user facilities for manipulating connector specifications.

5.1 Syntax

We take the following JTom code where the module Connector from the construction \%gom{...} specifies the BNF grammar of connectors:

```java
import testconnector.connector.types.*;
public class TestConnector{
   %gom{
      module Connector
         imports Stream Logic
         abstract syntax
            Node = node(name:String, s:StreamId) | source(n:Node) | sink(n:Node)
            Ins = ins(Node*)
            Outs = outs(Node*)
            Config = R(ins:Ins, outs:Outs)
            ChannelType = sync() | fifo() | syncDrain() | lossysync() | ...
            TChannelType = timer() | offtimer() | resettimer() | ...
            Pre = pre(p:Pred)
            Post = post(p:Pred)
            CSpec = spec(p:Pre, q:Post)
            constraint(t:ChannelType, s1:StreamId, s2:StreamId)
            Connector = channel(ct:ChannelType, n1:Node, n2:Node)
            tchannel(ct:TChannelType, time:Double, n1:Node, n2:Node)
            connector(cf:Config, cs:CSpec)
            ConnectorList = connectorList(Connector*)
   }
   public final static void main(String[] args) {
      Connector ab;
      StreamId sa, sb;
      sa = 'sId("sa")'; sb = 'sId("sb")';
      ab = tchannel(timer(), 5, source(node("a", sa)), sink(node("b", sb)));
   }
```

In order to use the Java code generated by the \%gom{...} construction we need to import the package testconnector.connector.types.* where the path corresponds to the name of the class (TestConnector), followed by the name of the module (Connector), both in lowercase, and ending with types.*. Furthermore, the module Connector imports the module Stream, where the BNF grammar of streams is defined as pairs of data and time sequences, and the module Logic, which defines predicates as basic blocks for building connector specification. These predicates include, besides the booleans True() and False(), WD for the well-definedness of streams and M for the merging operation. The \%gom{...} construct defines sorts for data-structures, e.g., Node, Connector, and operators on these sorts. The operators node, connector, channel and tchannel are called constructors in an algebraic sense, because they are used to “construct” connectors either from configurations and specifications or from nodes, in the case of basic (untimed or timed) channels.

The main method illustrates the declaration and definition of connectors and streams. The definitions use the back-quote expression ` to construct data. The operator sId is declared in the module Stream to identify streams. Note that we work symbolically with identifiers instead of real values because we do not actually need them. As we will see later, whenever the identifier of a data or time sequence is needed, we only need to
call a corresponding function which unfolds streams. Note further the use of the constructor \texttt{tchannel} for defining \texttt{ab} as a basic channel of type \texttt{timer()} that has a source node \texttt{a} with a stream \texttt{sa} and a sink node \texttt{b} with a stream \texttt{sb} where the delay is 5 time units. We distinguish between the constructors \texttt{connector}, \texttt{channel} and \texttt{tchannel} because we do not want users to explicitly write the specification of basic channels; to define a channel, one only needs to specify the type and the nodes of the channel. And for timer channels additional timing information needs to be provided, this is what the definition \texttt{ab} illustrates.

5.2 Semantics

We now describe how the semantics of connector designs is implemented in JTom. We first consider the specification of connectors. Connectors are either basic (untimed or timed) channels, or have the generic format of pairs of configurations and specifications. To extract the specification we define a \texttt{Java} function which uses one of the main features of \texttt{Tom}, the construction \%match:

\begin{verbatim}
public static CSpec getSpec(Connector c) {
  %match(c) {
    connector(_, spec) -> { return 'spec; }
    tchannel(timer(), 5, source(_, sa), sink(_, sb)) &&
       comp(da, ta) << getUnfold(sa) && comp(db, tb) << getUnfold(sb)
      -> { return 'spec(pre(andL(WD(comp(da, ta)), leT(add(ta,5), tail(ta)))),
                       post(andL(WD(comp(db, tb)), eqT(add(ta,5), tb), eqD(timeout, db)))); } } }
\end{verbatim}

where “\_” denotes an arbitrary variable and can be used when the name of the variable is not needed. The \%match construct is similar to a \texttt{switch/case} mechanism in imperative languages or to a rule in declarative languages: given a subject, if it matches a pattern, the associated action is executed. For example, when the subject \texttt{c} matches the pattern \texttt{connector(_, spec)} denoting a connector in the general format, the associated action is to return the specification of \texttt{c}. Besides patterns, \texttt{Tom} allows the definition of constraints. For example, the expression \texttt{comp(da, ta) \&\& getUnfold(sa)} denotes a constraint between the pattern \texttt{comp(da, ta)} and the subject \texttt{getUnfold(sa)}. It basically says that the stream identified by \texttt{sa} has a data (time) sequence identified by \texttt{da} (\texttt{ta}). The boolean operator \&\& is used to combine multiple constraints. Thus, when the subject \texttt{c} matches \texttt{tchannel(timer(), 5, source(_, sa), sink(_, sb))} which denotes a \texttt{Timer} channel with a stream \texttt{sa (sb)} on its source (sink) node with delay of 5 time units, and the constraints on the streams are satisfied, then the action returns the specification of the \texttt{Timer} channel, i.e., the precondition of the well-definedness of \texttt{sa} and the post-condition of the well-definedness of \texttt{sb}, together with the constraints that the data (time) sequences should be satisfied. Observe that the advantage of the \texttt{Tom} syntax is its modularity and expressiveness. This leads to more clear and concise code.

The next step in implementing the semantics of Reo is to consider the composition operators, i.e., sequencing, merging and replicating. In our semantics of Reo, merging and replicating have a higher priority than sequencing, so we need to be able to specify that sequential composition takes place after all the other possible compositions have been performed. Since we consider connector compositions as elementary transformations, we implement them as elementary strategies by means of the \texttt{Tom} construct
In this way, we maintain a clear separation between transformation and control. Strategies also afford us a great degree of flexibility and this makes it possible to experiment with different alternative scheduling policies for compositions, by choosing alternative definitions of strategies. Due to the length limitation, we do not describe the strategies in detail here. To understand the idea it suffices to know that their corresponding strategies are `Seq()`, `Replicate()` and `Merge()`. Taking into account that merging and replicating have a higher priority than sequencing, we implement a function that takes a list of connectors as input and returns its corresponding “normalised” connector, obtained by applying all possible compositions:

```java
public static ConnectorList fixpoint(ConnectorList cl) { 
    try {
        cl = `Repeat(Choice(Merge(), Replicate())).visit(cl);
        cl = `Repeat(Seq()).visit(cl);
    } catch (VisitFailure e) { System.out.println("fixpoint: strategy failed"); }
    return cl; }
```

In the above code, `Choice` is a basic strategy combinator with intuitive meaning and `visit` is the method for applying a strategy to a term, in our case to the list of connectors `cl`, and the `fixpoint` function relieves the programmer from the burden of explicitly specifying the order of compositions.

## 6 Conclusion and Future Work

This paper extends our previous work on the design model for (untimed) Reo connectors and introduces the UTP design model for timed Reo connectors. This approach provides a unified semantic model for different kinds of channels and composite connectors, covers different communication mechanisms encoded in Reo, and allows the combination of different untimed and timed channels as in Reo. In our work, we model basic channels in Reo as designs in UTP, and the composition of connectors is specified by design composition. Our semantic model offers potential benefits in developing tool support for Reo. The syntax and design semantics for timed Reo connectors are implemented in Tom.

In future work, we will investigate the semantic model of probabilistic connectors [7]. On the other hand, we will investigate the relationship between the UTP design semantics and other semantics of Reo that have been developed, and extend the UTP design model to treat the inherent dynamic topology and mobility in “full” Reo, especially context-sensitive connector behavior and reconfiguration of connectors. The development of refinement and testing theories for timed connectors like refinement and testing for untimed connectors in [3,16] and integration of such theories into the existing tools for Reo [1] is of special interest and in our scope as well.

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A New Component Model for Highly Distributed Environments

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1 Abstract

During the last years new distributed platforms have emerged, often qualified as highly distributed environments (HDEs). HDEs still include powerful and robust machines but they are, in addition, composed of resource-constrained and mobile devices such as laptops, tablet computers, personal digital assistants (or PDAs), smart-phones, GPS devices, sensors, etc. Moreover, these devices communicate using a variety of dependable and undependable fixed and wireless networks.

This fundamental change in the deployment environment has not been accompanied by a software model that provides deep understanding and systematic solutions to build compatible software systems. This challenge has been reported as a problem by software developers, software architects, and computer science researchers. Mark Murphy said: “Developing programs for a phone is a different experience than developing desktop applications, web sites, or back-end server processes” [6]. Similarly, however on a different level of abstraction, Malek et al. [5] have noticed “transparency (i.e. hiding distribution, location, and interaction of distributed objects) is considered a fundamental concept of engineering distributed software systems, as it allows for the management of complexity associated with the development of such systems”. This is usually achieved through the utilization of a middleware layer that has as main function (among others) to make remote calls appear as local calls. That is correct for stable distributed systems, however, this same concept, distribution transparency, has been shown to suffer from major shortcomings when applied extensively in HDEs [5]. Similar arguments have been given by Luca Cardelli [3].

That leaves us in the following situation: there is excessive and increasing need to build complex mobile and pervasive systems for entertainment and professional uses. And at the same time, the fundamental engineering techniques available are inherited from stable distributed environments, and suffer from several drawbacks and weaknesses when utilized in these new environments. The only available answer currently is applying ad-hoc techniques to overcome these drawbacks and weaknesses.

This ongoing Ph.D. work is a direct response to the above mentioned challenge. We are trying to provide a comprehensive answer for the above mentioned challenge. The solution we propose is based on a “paradigm shift” from distribution transparency to localization acknowledgment. Traditional distributed applications are based on distribution transparency, where a middleware layer is
expected to handle and hide all remote communications. Moreover, distribution transparency also masks many distinctions between devices, such as processor architecture and operating systems, by utilizing other software pieces including but not limited to Java virtual machine. On the other hand, localization acknowledgment removes these masks completely and to the deepest level needed. All details of the deployment environment, including networks and communications, mobile devices, constrained devices, sensors, operating systems, protocols, and even algorithms inside protocols and operating systems, are considered a first class concern in this new model that we propose in this PhD work.

To facilitate this paradigm shift in an implementable approach, we used the concept of “software components” because of its attractive and powerful encapsulation attributes. We propose a novel component model called Cloud Component (CC). This model includes the expected deployment environment in its definition, i.e. we raise the importance of the deployment environment to be equal to the importance of the functionality required from the component, with contrast to the other component models that are centered over the required functionality only [4]. The other important feature of this novel model is that it is fundamentally distributed. A single CC is usually distributed over many distant hosts, the specification of these hosts are considered and fundamentally acknowledged during the development process of this CC, and all aspects related to communication, coordination, and quality of service (QoS) monitoring and maintenance, are migrated to be internal to the border of the CC. This makes CC novel and nontraditional with comparison to traditional component models that, in most models, provide no support for extra functional properties (EFPs) such as QoS [4]. This is particularly true for industrial component models such as OSGi and EJB. And the fewer models that provide support for EFPs fails at the composition of EFPs. “Clearly, the composition of EFPs still belongs to the research challenges” [4].

The novel development process we propose starts with required specifications of the application along with the expected deployment environment(s). All steps of the development process are directly guided by the specification of the deployment environment devices, networks, etc. We also proposed an assembly model to build large systems using CCs as building blocks. To support software development using CCs, we provide automatic checkers to increase the level of confidence in our software. The first checker is the assembly checker that verifies if the CC assembly in the design respects the guidelines suggested in the CC assembly model. This checker generates warning and error messages to help the designer(s). The second checker is the software-hardware compatibility checker. This checker is an ontology-based checker [1, 2]. Before deployment, we model the actual deployment environment where we are going to deploy the system. An automatic checker will verify if this actual deployment environment is compatible with the expected deployment environment that the system was designed and implemented for. If yes, the deployment proceeds with high level of confidence of QoS. The second checker is partially implemented. We also proposed a formal notation to describe CC model, CC assembly, and the development process of
CCs. We believe this formal language will open the door for a wide range of theoretical topics including component type inference, subtypes, etc. Moreover, this formal language is a fundamental tool to define, prove, and communicate concepts related to CC model. The above mentioned checkers are based on this formal model. Due to the lack of space, we will present briefly the formal definition of the CC, leaving all other definitions and formalism for future publication.

A single cloud component is formally defined using the following four tuples: (1) A finite set of roles \( \mathcal{A} \). (2) A finite set of multiplicities for these roles \( \mathcal{M} \). (3) A set of possible deployment environments \( \mathcal{L} \). Each \( L \) is either a finite set of hosts \( \mathcal{H} \), or a finite set of host types \( \mathcal{T} \). (4) A function \( Z \) that maps roles to location types or hosts.

\[
\Omega = (\mathcal{A}, \mathcal{M}, \mathcal{L}, Z)
\]

The following formally defines the cloud component \( com \) in figure 1: \( \Omega com = (\mathcal{A}, \mathcal{M}, \mathcal{L}, Z) \) where: \( \mathcal{A} = \{ AS, AR \} \), \( \mathcal{M} = \{ (AS, 1), (AR, 2) \} \), \( \mathcal{L} = \{ \{ TServer, TClient_1, TClient_2 \} \} \), \( Z : AS \downarrow TServer \), \( AR \downarrow TClient_1 \), \( AR \downarrow TClient_2 \).

This is an ongoing work. At the current stage we are developing real scale software application using CC as building blocks, and using all the above mentioned tools and ideas. We will use this software application to refine our proposal, and finally as a prove of concept.

References

A Rewriting-Logic-Based Tool for Object-Oriented Formal Modeling and Analysis of Interacting Hybrid Systems

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Introduction. HI-Maude [4] is a tool that supports the formal modeling, simulation, and model checking of interacting hybrid systems in rewriting logic. The tool is targeted for complex hybrid systems in which multiple physical entities interact and influence each other’s continuous and discrete behaviors. For example, a cup of coffee in a room interacts with the room and a coffee heater through different kinds of heat transfer (see Fig. 1, right), leading to a change of the coffee’s temperature, and this change also influences the change of the room’s temperature. More heat flow from the heater may cause the coffee to evaporate, leading the phase change of the coffee from liquid to gas. One distinguishing feature of HI-Maude is the modularity and compositionality of the specification of the system’s continuous dynamics. Non-compositional specification of the whole system is very hard, as it involves combining the ordinary differential equations (ODEs) that specify the dynamics of its components; it also requires redefining the system’s continuous dynamics for each new configuration of interacting physical components.

Modeling and Executing Interacting Hybrid Systems. We have defined a general object-oriented modeling methodology [1] for modeling such interacting hybrid systems in HI-Maude, which adapts the effort/flow method [7] to model a physical system as a network of physical entities and physical interactions between the entities. This makes the models modular and compositional, in the sense that it is sufficient to define the continuous dynamics for each component to define the dynamics of the entire system. To model such interacting system we have defined three kind of components (see Fig. 1, left): (i) a physical entity is described by a real-valued effort, a set of attribute values, and the entity’s continuous dynamics; the attribute values describe discrete properties, e.g., the mass or the phase of a material, that can only be changed by discrete events; the effort variable represents a dynamic physical quantity, such as temperature, that evolves over time as given by the continuous dynamics in the form of an ODE (ii) a physical interaction between two physical entities is described by a real-valued flow, a set of attribute values, and a continuous dynamics; the flow value describes the dynamic interaction between two entities, whose evolution over time is specified by the continuous dynamics; the values of the effort variables of the two physical entities are used in the definition of the continuous dynamics of the interaction (iii) a flow source can be use to model the addition or removal energy from a physical entity.
The system components may also exhibit discrete transitions, because of phase changes, explicit control, communication, or other factors.

We use numerical techniques to approximate the continuous behaviors by advancing time in small discrete time increments, and approximating the values of the continuous variables at each “visited” point in time. We have adapted the following numerical methods to our effort/flow framework: the Euler [1], the Runge-Kutta 2nd order, and the Runge-Kutta 4th order methods [3] for fixed and adaptive time increments. Another method which will be adapted to the tool is the approximation-error-based adaptive time increments which its Real-Time Maude implementation is presented in [2] (adapting the Runge-Kutta-Fehlberg 4/5).

The HI-Maude Tool. The HI-Maude tool extends Real-Time Maude [6], which is a formal specification language and a simulation, reachability analysis, and LTL and TCTL model checking tool for real-time systems. It is based on rewriting logic [5] and emphasizes expressiveness and ease of specification.

The HI-Maude tool integrates the modeling techniques explained above and the Real-Time Maude implementations of the adaptation of the numerical approximation algorithms to support the rewriting-logic-based object-oriented formal modeling and simulation, reachability, and LTL model checking analysis of hybrid systems containing interacting physical components. HI-Maude extends Real-Time Maude’s analysis commands by allowing the user to select (i) the numerical approximation technique used to approximate the continuous behaviors, (ii) the time increment used in the approximation, and (iii) discrete-switch-detection-based adaptive time increments.

In particular, the HI-Maude tool makes it easy to define the continuous dynamics of the effort and flow variables of single physical entities and physical interactions, respectively. Once the dynamics of the single physical components have been defined, the tool (i) automatically defines the continuous dynamics of the entire systems, and (ii) provides the usual Real-Time Maude formal analysis commands, but where the desired built-in approximation algorithm, the desired time increments used by the approximations, and the activation of discrete-switch-based adaptive time increments are additional parameters of the
commands. Furthermore, the tool provides infrastructure to define that instantaneous transitions (modeled as instantaneous rewrite rules) are applied in a timely manner.

HI-Maude is as expressive as simulation tools, yet provides reachability and LTL model checking analysis in addition to simulation. The price to pay is that reachability and satisfaction of LTL properties are in general no longer decidable. HI-Maude only analyzes those behaviors that are possible with the selected time increment and numerical method used to approximate the continuous behaviors. Therefore, the results of search and model checking in HI-Maude may not be correct. If a counterexample is found in LTL model checking, or a desired state is found in a search, these are indeed valid counterexamples up to the approximation errors due to the use of numerical approximations and round-off errors due to the use of floating-point numbers. However, since only a subset of all possible behaviors are analyzed, the fact that a state is not found during a search or that LTL model checking returns true does not necessarily imply that the state cannot be reached or that the LTL property holds.

Currently we are working on a complex case study on modeling human thermoregulatory system. We have made many improvements to the human body model presented in [4]. We are targeting to model and analyze problems related to heat and cold exposures to human body by using real examples, e.g., the problems in sauna bathing, winter activities, firefighting, etc. We compose the human body model with different models representing different kinds of environment surrounding human body.

The HI-Maude tool files, including case studies and related papers and technical reports, are available at http://folk.uio.no/mohamf/Hi-Maude/.

References

Formal Aspects of Component-Based Design of Embedded Real-Time Systems

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Motivation and Previous Work

An embedded system is traditionally defined as a computational system that is a part of a larger device and serves a specific function. Such systems are often safety-critical, and a vast majority of them include at least some hard real-time tasks, where correctness depends on timeliness of computations.

In recent years, there has been a surge in complexity of embedded systems, often further complicated by resource sharing between independent tasks; there is also a growing demand for concurrency in system operation. The challenge is to achieve that without compromising correctness and reliability.

Component-based design (CBD), successfully used in development of general-purpose software, seems to be a natural choice. However, despite a considerable body of research (see, for example, work on Rubus Component Model [2] and Ptolemy II [1]) and a number of commercial tools available in the market, CBD is rarely used for development of embedded systems outside some specific domains (e.g., Koala [3] has been used by Philips in consumer electronics). This calls for a new, formal approach to CBD addressing the important characteristics of embedded systems. Such an approach should:

– allow for an accurate modeling of interaction between software and hardware, which is key to defining functionality of most embedded systems and is the usual source of hard real-time requirements,
– incorporate timing specifications, central to development of most embedded systems, directly into the model at all levels,
– provide an intuitive and “safe” mechanism for introducing concurrency into system operation,
– ensure a good correspondence between a component model and its implementation in a programming language, in order to facilitate component and system verification, and
– define a graphical representation of a component model of a system that can be used throughout the design process (in most tools in use today, a graphical model can no longer be used once program code generated from it is modified manually).

Our work on CBD of embedded real-time systems is a part of the ongoing development of the programming and modeling language Timber [4] at Luleå
University of Technology and Chalmers University of Technology, Sweden. Timber specifically targets hard real-time systems. It is an object-oriented language that combines purely functional evaluation of expressions with an imperative-style command layer, incorporating system I/O and message passing between objects. Importantly, it has been developed to meet some of the requirements on modeling embedded systems discussed above. Firstly, it adopts a reactive execution model, i.e. a reaction (execution of an object’s method) is triggered either by an external event or by a message from another object; this makes Timber particularly suitable for specifying interaction between software and hardware. Secondly, it combines object-level concurrency with a complete encapsulation of an object’s state, which results in a fairly simple implicit concurrency model. Thirdly, timing specification can be incorporated into Timber code by specifying a \textit{permissible execution window} – a baseline and a deadline – for each method invocation; this specification is preserved in the compiled code and can be used to guide scheduling at run-time as well as to perform static schedulability analysis.

**Research Goals**

Our goal is to formulate a general framework for component-based design of embedded real-time systems and to include support for CBD directly in the Timber language. This involves a number of tasks:

- a formal component model should be defined using existing syntactic constructs or by introducing new ones to the Timber language, with component composability verified by static type-checking (Timber has an advanced type system with automatic type inference); the unambiguous semantics of such a model is guaranteed by the operational semantics of the language;

- a language for specifying timing behavior at system and component levels should be defined so that it is possible to verify (a) component composability with respect to timing requirements, and (b) that a particular implementation of a component (in terms of Timber code with permissible execution windows for method invocations) conforms to timing specification at the component level;

- a graphical representation of a component model should be defined, as well as its translation to Timber code and vice versa; the existence and quality of this representation is seen as key to usability of the proposed approach.

Some of our recent results obtained while working on the third task are presented below. These results are of a general nature and are not specific to Timber-based component models.

**A Formal Approach to Defining a Graphical Representation**

A formal component model is typically defined in a text-based language (in our case – in the programming language Timber), but its usefulness depends on the existence of a graphical representation. True usability, however, comes with the
ability to perform graphical editing operations at any stage of the development process, i.e. even after program code has been added and/or edited manually (which is not supported by the majority of existing design tools). The challenge here is two-fold: firstly, information that should be visualized in a graphical model is often implicit in program code (e.g., a relation between two components can hinge on the value of some variable), and secondly, a graphical model should be simple and hence the translation from program code is inevitably lossy and in many cases irreversible.

To meet this challenge, we propose to introduce an intermediate representation $G$ and, given a set of abstract syntax trees of syntactically correct programs $P$ and a set of models $\hat{G}$, expressed as labelled graphs (such as formal component models or any other models with unambiguous graphical representations), we define:

- a translation $\tau : P \rightarrow G$ that augments an abstract syntax tree of a program with an explicit representation of information needed for the graphical model (e.g., links between a component definition and its instantiations); we should require that for every syntactically correct program there exists exactly one intermediate representation;
- an inverse translation $\tau^{-1} : G \rightarrow P$ that removes the vertices and edges added by $\tau$ (these may be labelled differently from labels on vertices and edges of abstract syntax trees to make $\tau^{-1}$ trivial), such that

$$(\forall p \in P)[\tau^{-1}(\tau(p)) = p]$$

- an “erasure” $\delta : G \rightarrow \hat{G}$ that hides some of the information available in $G$ (e.g., pure computations that do not affect the component structure) by “collapsing” certain vertices into edges and by completely removing some other vertices and edges;
- graphical editing operations $\gamma_1, \ldots, \gamma_n$, defined not on $\hat{G}$ but on the more complete intermediate representation $G$; for each operation it is necessary to prove that it preserves syntactical correctness of a program:

$$(\forall p \in P)(\exists p' \in P)[\tau^{-1}(\gamma_i(\tau(p))) = p']$$

It can be shown that this approach ensures that a graphical representation can indeed be used at any stage of the development process, even after program code has been edited manually.

References

Analysis of Cooperating Systems by Refined Over-Approximations

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We consider complex systems which are built from cooperating subsystems (components). The reachable state space of a system can become exponentially large in the number of components. For many formalisms, modeling this kind of systems, there exist PSPACE-completeness results for the reachability problem – even for classes of systems with strong structural constraints. Therefore, many important properties of such systems can not be checked efficiently.

Here, we introduce an approach for verifying a global property \( P \) in a complex cooperating system. The approach consists of three parts:

1. the construction of a family of compact representations of over-approximations of the global system,
2. the refinement of these over-approximations by a technique that we call Edge-Match and
3. the construction of an indicator predicate \( P' \), such that \( P \) holds for a system under consideration if \( P' \) holds for all over-approximations.

Our approach has polynomial costs, i.e., we can not expect that our approach works for every system.

We consider the component-based formalism of interaction systems (Sifakis et al.) as a formal model. A tuple \( Sys = (K, \{A_i\}_{i \in K}, Int, \{T_i\}_{i \in K}) \) is called interaction system. \( K \) is a set of components and, for \( i \in K \), \( A_i \) is a set of ports and \( T_i \) a transition system with labels in \( A_i \) that models the behavior of component \( i \). The cooperation of the components is specified by a set \( Int \) of interactions. An interaction is a set of ports from different components, which have to work synchronously. The behavior of \( Sys \) is a transition system \( T \). The state space of \( T \) is the Cartesian product of the local state spaces of the \( T_i \). The transitions are specified by \( Int \). A transition with label \( \alpha \in Int \) can be performed if for each involved component its participating port is enabled.

The over-approximations are constructed by choosing a subset \( C \subset K \) of components and constraining the interactions \( Int \) to these components. An interaction system \( Sys_C \) that is defined in this way can be interpreted as a compact representation of an over-approximation that would result, if the components in \( K \setminus C \) have always all their ports enabled. Thus, if a state \( q \) is reachable in the global behavior \( T \) then a state \( q_C \) is reachable in the behavior of \( Sys_C \) such that \( q_C \) is a projection of \( q \) on the components in \( C \).

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An over-approximation that is constructed in this manner might be relatively coarse. In a second step we refine the approximations by an operator we call Edge-Match that compares the $Sys_C$ pairwise. The operator is based on the following observation. Let $C, D$ be subsets of $K$ such that $C \cap D \neq \emptyset$. Let $q_C \xrightarrow{\alpha} q'_C$ be a reachable transition in an over-approximation with respect to $C$. If there is an over-approximation with respect to $D$ with no reachable transition $q_D \xrightarrow{\alpha} q'_D$ where $q_C$ and $q_D$ respectively $q'_C$ and $q'_D$ agree on shared components then there is no reachable transition in the global behavior $T$ that is a projection of $q_C \xrightarrow{\alpha} q'_C$. Thus, removing $q_C \xrightarrow{\alpha} q'_C$ results in a refined over-approximation with respect to $C$.

We demonstrate by the property of deadlock-freedom, how the over-approximations can be used to verify a property $P$. A deadlock in $Sys$ is a state in $T$ in which no interaction is enabled. $Sys$ is called deadlock free, if no deadlock is reachable in $T$. In a deadlock every component is waiting (in its local state) for other components to synchronize its enabled ports. This implies the existence of a directed cycle of components that are waiting for each other. If a deadlock is reachable in $Sys$ then this cycle is visible as chains of waiting components in according approximations. A (naive) indicator predicate $P'$ can be stated by: there are no such chains in a (reasonable) set of approximations. If $P'$ holds then $Sys$ has no reachable deadlock. We use a sophisticated version of this predicate that consists amongst others of a second application of a modified form of the Edge-Match operator.

Table 1 displays results of our approach for the refinement of over-approximations on a couple of parameterized examples. Here we used over-approximations with respect to all subsets that consists of three components that have a connected communication structure – this family of over-approximations is suited to our indicator predicate for deadlock freedom and depending on the structure, this family can be notably smaller than a family built from all subsets of size three. The table reads as follows: $|K|$ is the number of components and $|Int|$ the number of interactions of the interaction system under consideration. $|Q|$ is the number of states of the corresponding global behavior, i.e., the number of states that have to be dealt with in a direct application of model-checking techniques. #OA is the number of over-approximations. #States is the sum of the size of the state spaces of the over-approximations. #RT is the sum of the number of reachable states in the initial over-approximations. #EM is the sum of the number of reachable states in a fixed-point with respect to reachability analyses and the application of the Edge-Match operator. “time in ms” is the time in milliseconds that was needed to calculate the fixed-point on a 2.53GHz dual core CPU with 4GiB memory. The results show that a significant number of states is removed by our operator which could be the reason for a false-negative of an indicator predicate. The results are calculated by our tool that is based on BDDs as a data structure for transition systems.

Our approach offers the possibility to effectively check properties of systems whose state space size is too large for a direct application of model-checking
techniques. We described how over-approximations can be used for checking the important property of deadlock-freedom.

Even, if our approach works in praxis efficiently, so far we can give only a rough polynomial upper bound for the calculation of a fixed-point with respect to our refinement operator. The Edge-Match operator is related to the Semi-Join operator in the relational data model. This relationship will be analyzed further to give amongst others a detailed upper bound for particular classes of systems. The main idea is that a transition relation of a transition system can be interpreted as a table in the relational data model, i.e., a transition in an over-approximation is a row in an associated table such that every state of a component and the interaction is placed in its own cell. To put it roughly, an application of the Semi-Join operator on two tables that model over-approximations results in tables of transitions that agree on shared components. An important theorem in the relational data model states that a fixed-point without “artifacts” with respect to the Semi-Join operator on a family of tables exists iff the hypergraph with respect to the attributes of the tables is acyclic. The number of Semi-Joins needed to calculate a fixed-point on a family like this is linear in the number of tables.

| System        | |K| | Int| | Q| #OA | #States | #RT | #EM | time in ms |
|---------------|---|---|----|----|-----|-----|--------|-----|-----|-----------|
| Circle(30,20) | 90 | 270 | 2428,3 | 240 | 5.790.240 | 4.519.260 | 3.152.640 | 59.799 |
| Chain(30,20)  | 88 | 372 | 2420,5 | 225 | 5.931.613 | 4.567.165 | 2.780.289 | 50.287 |
| Grid(10.5,3)  | 165 | 700 | 2486,6 | 2.151 | 9.149.931 | 7.297.031 | 6.383.026 | 99.729 |
| Phil(500)     | 1.000 | 2.000 | 2792,5 | 1.000 | 48.000 | 25.500 | 20.000 | 11.334 |
| PhilTB(200)   | 600 | 800 | 2864,4 | 7.800 | 144.000 | 110.000 | 69.200 | 256.390 |

Table 1. Results of a couple of parameterized examples. Underlined parameters affect the size of components, non-underlined the number of components.

\(^1\) a) Circle - a circle of components, b) Chain - a chain of components, c) Grid - a rectangular grid of measurement stations, d) Phil - classic philosophers problem, e) PhilTB - classic philosophers problem with semaphores