THE CLIENT/SERVER MODEL IN DISTRIBUTED COMPUTING

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Department of Computer Science
University of Utrecht
P.O.Box 80.012, 3508 TA Utrecht
The Netherlands
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Abstract

A variety of design issues for distributed computer systems is explored in terms of the Client/Server model. An attempt is made to expose the basic ingredients of the Client/Server model, in order to facilitate further theoretical analyses and applications in the design of complex systems. In addition, a detailed 2-phase commit protocol is presented and verified in a form that is suitable for applications in distributed systems.

Keywords and phrases: distributed systems, computer networks, remote operations, client/server model, message passing, atomic actions, two-phase commit protocol.

1 Introduction

Increasingly computer networks and distributed systems are being introduced in traditional application environments for computers. Both local and wide-area networks and their supporting software are being produced to meet with customer demands, and a considerable research effort is invested worldwide to solve the new and complex design issues that come with the development of distributed information systems. In this paper we will explore remote operations architectures, which allow for flexible client/server type interactions between distributed application processes on top of an existing and reliable transport level. In such architectures sets of primitives are provided for application association and message passing following a request/reply protocol, with facilities for service authorization and recovery. The remote operations architectures usually conform to internationally agreed standards, for the general requirements of interconnection and interchangeability.

In order to deal with the complex issues of communication and control in a distributed system, it is necessary to have a consistent architectural model underlying the design and development of a system. Additional requirements are imposed by the agreed ISO/ECMA standards for remote operations and transaction processing. In many distributed systems the client/server paradigm is used to explain the underlying system view, suggesting the possibility of a formal model and correctness proofs of the design. In this paper an attempt is made to expose the essential ingredients of the Client/Server model which should be taken into account, in a form that should facilitate the use and further study of the model.

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for both practical and theoretical purposes. We assume some familiarity with the area of computer networks (see e.g. Tanenbaum [30]).

The Client/Server model can be recognized in many distributed system architectures. Without much difficulty all aspects of distributed system design at the network level and beyond can be brought to bear on the model. Yet, despite its ubiquity, no detailed study seems to have appeared that presents a complete review of the Client/Server model. This paper is a modest attempt to fill this gap. We will address a large number of issues that are not well-documented in the current literature nor analysed very well theoretically, and the paper hopefully serves as a stimulus for further research. We do not develop a prototype model or design in ISO/ECMA frameworks here, but will outline a number of ideas that might enhance the Client/Server model as a basis for distributed system design.

A remote operations system comes with two design aspects: an "applications architecture" describes the system from the perspective of an applications programmer, and a "communications architecture" describes the available network functions. We adopt the view that an application may consist of several, distributed application processes that cooperate to achieve some goal. An application may exist "forever", i.e., may have to be supported for an indefinite period of time. An application (i.e., an application process) will be activated by an external cause or by another application. Application processes can activate network services that are generally available. Network services in turn can activate other network support tasks. In addition, autonomous processes may exist that play a role in standard network control and function maintenance.

Application processes communicate using the facilities of the communications architecture. To shield application processes from the details of the functions provided by the communications architecture, one or more interface modules can be designed that provide "higher lever" communications facilities.

In this paper we will describe the Client/Server model as a unifying framework for the high level description and specification of communications and interactions between processes. The model is geared to understanding general Client/Server architectures, but carries the idea of the model much further. The model as we describe it only serves as a structuring tool for a network architecture at the session and presentation layer level. The model is heavily based on the "abstract object" approach to distributed system design (see e.g. Watson [33]) and aims at viewing a number of aspects of Client/Server architectures from this perspective. The abstract object approach has successfully been applied in other contexts of networking (see Sloman & Kramer [27] for an extensive summary).

The paper is organized as follows. In section 2 we will briefly introduce the issue of distributed computing, the Client/Server model and the notion of Remote Function Modules. In section 3 the Client/Server model is presented in more detail. In section 4 a large number of architectural issues in computer networks are reviewed from the unifying viewpoint of the Client/Server model. In section 5 we present some ideas for implementing the Client/Server model in general environments, using the notion of Remote Function Modules. Section 6 discusses the complex issues concerning binding and unbinding. Finally, in section 7 we consider more involved "atomic" interactions between clients and servers and the need for atomic commit protocols. An integral development and correctness proof is given for a standard 2-phase commit protocol that seems to be appreciably simpler and more clarifying than similar treatments in the current literature (see e.g. [2,6]).
2 Distributed Systems and Remote Operations

A large number of computer-oriented hardware or software systems are referred to as "distributed systems". Generally, the phrase is used to refer to any network of communicating machines or processes designed to achieve some overall goal of a collection of users or an organisation. It can refer to the supporting network hardware and software only (which is a very common use of the term), or to a distribution of application programs and control functions over various sites with an underlying scheme for concurrent operations and communication. In all cases there are complex issues of system operation and control not found in traditional computer applications, and encountered in only a limited form in the modern approaches to operating systems (see e.g. Peterson & Silberschatz [25]). There are a number of different approaches that can be recognized in the design of distributed systems:

(a) the Hierarchical model, and

(b) the Client/Server model.

In the Hierarchical model, an executing program can send subtasks to other machines for execution. The model is suitable for applications that have a hierarchical structure and is commonly found, for example, in distributed database systems (see e.g. [6]). The model requires the supporting network to provide a service like "send subtask X to another machine (or, to machine Y)" and some form of message passing between a program and its subtasks. In some cases it is possible to send the complete code of a subtask to another machine. More often it is possible to "call" a stored copy of a program at the other machine and pass it the necessary parameters and/or data in some form or another, like in systems that implement remote procedure calls (RPC) as described in Birrell & Nelson[6]. Normally program execution is not resumed until the result of a subtask has been received.

In the Client/Server model, each program may request services from designated service providers (servers) that are shared with other users or application programs and that are usually situated on other machines. Examples of servers are: name servers, file servers, printer servers and other special programs. Again the model requires the network to provide a suitable interface for expressing and transporting service requests, and it is common to use the logistics of remote procedure calls or a relaxed, asynchronous version of it. In particular, a server call should not keep the calling program from further execution.

From a systems point of view the Client/Server model subsumes the Hierarchical model, although the requirements of control will be more complex. But in many respects the Client/Server model seems to be the more fundamental paradigm in distributed computing, as it appears at the basis of the network developments of major vendors like IBM[11] and NCR.

The Client/Server model is based on the idea that in a distributed system environment certain processes (clients) will be requesting functions from other processes (servers) and that all distributed activity can be seen from this perspective. The model tends to declare fixed processes (processors) as clients and others as servers, which will be adequate in systems where Client/Server functions are logically and physically separated. This clearly need not always be the case.

An obvious analogy with known notions from programming suggests to view Client/Server functions as logical entities that could well reside in a single process. More precisely,
we suggest to elevate the notion of Client/Server to the notion of a Remote Function Module (RFM).

The RFM notion is primarily a tool for structuring the specification, and hence the coding, of Client/Server applications. Remote function modules provide sets of functions that can be naturally grouped together, for local or remote use. Remote function modules can call on the services of other remote function modules.

Remote function modules facilitate abstraction, in the sense that its services can be used without necessarily understanding the way they are coded. When remote function modules are designed, irrelevant details can be suppressed by making use of other (lower level) remote function modules. Thus remote function modules facilitate hierarchies of abstractions, a known and powerful tool in system design.

In this paper the idea of remote function modules will be developed with three goals in mind:

(i) local/remote transparency (local and remote access should appear the same),
(ii) extensibility (it must be possible to add remote function modules to an application environment without great difficulty),
(iii) system structuring (desired services should be expressible in terms of remote function modules).

Remote function modules are not necessarily coded the way they are described here. RFM’s are primarily meant to structure, and do not necessarily guarantee efficient code in the form suggested.

3 The Client/Server Model

Before embarking on RFM’s, it is useful to consider some further aspects of the Client/Server model. The rationale for the Client/Server (or Customer/Server) model was first described by Gentleman [9], who identified the issues that need to be resolved in any system designed from this perspective. The Client/Server paradigm is now standard in distributed system design (see Comer [7]).

The Client/Server model can be viewed as a model for process interaction. One process (A), identified as the client, requests a service from some other process (B), which acts as the server. The server, after performing the requested service, posts a reply to the client. The client-server relationship between A and B exists only for the duration of the interaction. Thus a process that acts as the client in one interaction may become the server in another, and vice versa. We will later allow that servers request services from other processes while processing a request and thus act as server and client simultaneously. The Client/Server model is recognised in distributed systems based on the remote procedure call (RPC) paradigm (see e.g. Lampson [20] and Birrell & Nelson [4]) or the nested transaction paradigm (Moss [23]).

3.1 Requests and replies

The request of a client (A) must be passed to the server (B) in a specified way. Ideally it should be immaterial to the model whether B is local to A or remote. The same formats and procedures should apply in either case. In the Client/Server model, requests and
replies are passed as messages. (See e.g. Gentleman [9] for an appraisal of the various semantic issues involved in message passing.) An important issue is whether the message passing primitives are “blocking” or not. For example, after a request has been sent out, the client may remain blocked until a reply has been received from the server. (It is sometimes called “connection-less interaction”, because the association between the client and the server only exists for the duration of the request.) In the non-blocking case, the client typically engages in a sequence of message exchanges with the server but does not need the reply to each individual request before a new request can be sent out. (It is sometimes called “connection-oriented interaction”, because the client and the server will normally commit a fair amount of resources to maintain the connection.) The Client/Server model requires that a “session” is opened and maintained, in order for A and B to interact in this way. Later in this paper there will be a detailed discussion of the problems of connection management. There are various alternatives for the client to observe replies in the case of non-blocking requests. (Essentially there are blocking and non-blocking variants for it too, see e.g. the excellent discussion in Svbodova et al. [29].)

3.2 Complications because of possible failures

The Client/Server model is deceptively simple when communication failures and other failures are not taken into consideration. There are various types of communication failure that can occur: messages may be lost or “stuck” (e.g. because of deadlock), messages may be undeliverable (e.g. because one or more communication lines are down), and processes may “die”. Failures complicate the simple interaction scheme of the Client/Server model, because failures directly affect the request/reply mechanism. In particular, the reply a client receives on a request should either be a valid reply from the server or an indication that no valid reply from the intended server can be obtained because of some (identifiable or non-identifiable) failure. In the Client/Server model, a client must receive a reply (and no more than one) to every request it sends out, and the replies received must correspond exactly to the “facts” for the associated requests. It follows that replies are not simple and must be “interpreted”. In the model developed in this paper, replies strictly follow the Client/Server paradigm.

3.3 The use of timers

Failures affect the Client/Server model in yet another way. In physical communications systems, it is common to guard for failures by the use of positive acknowledgements and timers. Acknowledgements can be merged in with replies, but timers are different entities. Svbodova et al. [29] discuss the difficulty of “timeout parameters” for programming at length, but there is no question that timers are often the simplest and most adequate solution to communications control problems in concrete systems. For example, timers are used in the implementation of reliable remote procedure calls (see e.g. Lampson [20] or Shrivastava & Panzieri [26]) and “crash control” in transaction management systems (see e.g. McKendry & Herlihy [22]). In typical distributed systems timers are available implicitly (at the client side) or explicitly (at the server side) for request-handling purposes and session maintenance. At the client side, a timeout value can be set with every request to indicate how long one is willing to wait for a reply. A timeout is triggered when no reply is received in the prescribed time-interval or when an exception has occurred in the underlying transport layer. At the server side, a timeout value can be set with every
session. In this case the timeout value indicates how long the server is willing to wait for "new" requests from one of its clients viz. from the particular session. (Although not usually specified, it is assumed that the corresponding timer is reset to the timeout value each time a new request in the session is received before timeout.) Multiple timers must be implemented explicitly by means of incremental timeout queues (as described in e.g. Tanenbaum [30]). In all cases timeouts are handled as interrupts, and some action must be specified when a timeout occurs. At the client side it typically involves "retrying" a request that was timed out, which leads to the problem of duplicate detection by the server. At the server side a session that was timed out, is to be closed explicitly. In the model described in this paper we will assume that timeout values can be specified for all requests. (Semantically the use of timers by a server is part of the session-management protocol and not of the Client/Server model.) The functionality of any Client/Server-based system should only depend in a monotone manner on the value of the timeout parameters.

3.4 Sharing and concurrency

An important aspect of the Client/Server model is the fact that servers can be shared. It implies that any number of clients can independently request the services of a same server process B, and (hence) some policy must be set for every server of how to handle "simultaneous requests". In case all client-server interaction is session-oriented (as in most Client/Server architectures), the policy must essentially indicate whether a server can support only one session at a time or multiple sessions simultaneously. In the Client/Server architecture this has led to the distinction between "single-threaded" and "replicated" servers, respectively. The effect of concurrent sessions must be equivalent to the effect of some serial ordering of the sessions, in order to guarantee that semantically each session can still be regarded as the unique association between one client and the server. It follows that a server can be of the "replicated" type only when its semantics allows for it. When server replicas do not share resources, the semantic constraints are trivially satisfied and the effect is that of many clients having their own copy of the server and (thus) having access to the same services individually. In the Client/Server architecture, a syntactic difference can be made between single-threaded and replicated servers at the level of the allowable primitives.

3.5 More complex atomic actions

Aside from "many-to-one" interactions, the Client/Server model may allow for "one-to-many" interactions between processes. In this case, a client process (A) requests the services of various servers and (hence) engages in multiple sessions simultaneously. It is the client's task to synchronize the various sessions, when necessary. Additional complications arise if the simultaneous requests a client A sends out are not independent and constitute "one" (intended) transaction. Typically, a request should only be processed if all requests that are part of the transaction can be (by the various servers). This leads to the well-known problems of transaction management (see e.g. Gray [10], Ceri & Pelagatti [6], or [12]), viz. transaction commitment and recovery protocols. Atomic actions will be discussed further in section 7.
4 Client/Server Architectures

We now elaborate on a number of architectural issues for the Client/Server model. While some terminology is taken from [18], the material in this section has no particular relation to it.

4.1 Common Client Modules

The environment for a Client/Server architecture is a distributed system of “service providers”, i.e., processors of a known functionality connected by a local area network. Application processes may invoke operations that are only available across the network, at a remote site. The invocation of a remote operation triggers an involved mechanism that is (or, should be) largely transparent to the application process, and whose sole purpose is to perform the desired operation reliably and return the result to the invoker. (Note that the “result” may be a diagnostic e.g. indicating that the desired operation cannot be performed at the present time.) The Client/Server architecture as proposed in [18] suggests that invocations must be handled by a Common Client Module, or CCM. The CCM provides the same support to all application processes running on a given processor. An application process can invoke a remote operation by passing a suitable “operation request block” to the CCM. In fact, the CCM may maintain a queue of outstanding operation requests for each (local) application process. An operation request block must have a well-known format, including fields for e.g.

(i) the logical name of the requesting application process (the client),
(ii) a unique identifier for the request (e.g. a sequence number),
(iii) the logical name of the desired remote operation,
(iv) class information about the desired remote operation,
(v) parameter values for the desired remote operation,
(vi) a pointer to the “operation result block”,
(vii) reserved fields for use by the CCM,
(viii) the size of the operation request block.

The operation result block corresponding to each operation request must have a well-known format as well. (It obviously is sufficient for an application process to pass a pointer to the appropriate operation request block when invoking a remote operation through the CCM.)

4.2 Binding

The CCM must see to it that each operation request is shipped across the network to the appropriate service provider, and that the result of the operation request is written back into the corresponding operation result block. In order to ship a request across, the CCM must “bind” the service provider. For now it is sufficient to view binding as the initiation of a session between the client and a server process (the server). Binding can be implicit
or explicit. Clearly explicit binding requires that the (logical) name of the intended server is known. The session between a client and a server process is identified by a unique session-id which exists for the duration of the association.

4.3 Primitives and data structures

It follows that the CCM is a rather complex support module, when considered in its most general form. The CCM must support e.g. the following primitives:

(i) Bind.req ("bind request"),
(ii) Bind.repl ("bind reply"),
(iii) Oper.req ("operation request"),
(iv) Oper.repl ("operation result"),
(v) Cancel.req ("cancel an outstanding operation request"),
(vi) Cancel.repl ("result of cancelling an outstanding operation request"),
(vii) Unbind.req ("termination request for a session"),
(viii) Unbind.repl ("termination reply").

In addition the CCM may provide several other primitives, e.g. for inspecting the status of a current session with non-blocking operation requests. The CCM maintains a session-table with information on all current (active) sessions, including e.g. the unique session-id of each session and various status indicators. The session-table is implicitly maintained by the bind and unbind primitives. We defer a detailed discussion of binding and unbinding to a later section. With each entry in the session-table the CCM may maintain a queue of outstanding operation requests. The queues are implicitly maintained by the Invoke and Oper.repl primitives. (Interrupts caused by time-outs on outstanding operation requests could also be handled through the Oper.repl primitive, or a variant of it.)

4.4 Formats and message passing

There are several further issues involved in the design of a CCM. In order to ship an operation request from a client to the intended server process, the CCM must generate a message (or packet) and some kind of “interchange” must take place that will guarantee that the message is delivered to the server (and, eventually, that the operation result is delivered back to the client). Gentleman [9] gives an insightful discussion of the design considerations for message formats. In the design of Client/Server architectures one may assume that the problem of message formats is solved at a lower "level" and that the request/reply formats are "well-known", relying on a standard transport level to handle variable-length messages. The message formats will be related to the formats of the operation request and operation result blocks in a simple manner (compare e.g. [11]). In order that a message can be sent to the intended server, it is normally required that the network address of the service provider and the “port number” of the intended server are provided with the message (together, of course, with the return address for the reply message). We defer a detailed discussion of naming and addressing to a later section. Normally, the address information is maintained in a "connection descriptor" stored with the unique session-id. Thus the necessary information for name-to-address binding must be set up as part of the Bind.req and Bind.repl primitives. It also follows that the CCM must possess (and maintain) a network directory with the “well-known” address and attribute information of the available service providers on the network, in order that it can set up associations. Note that at the application (i.e., client) level, operations and servers need
only be known by generic names or aliases. In many virtually all distributed systems, a client process must name the intended server when invoking a (remote) operation.

Service providers (servers) may be invoked by remote clients to perform certain operations. In most distributed systems, servers must post their availability to the network as part of a "start up" procedure at initialization time. Typically this is achieved by "opening" the well-known socket for the server and authorizing the connection. In general servers "export" their availability and various other attributes (like the operations they provide) across the network and potential clients "import" this information into their network directory. Alternatively, it may be assumed that the servers on the network are "known", and the only status information that need to be maintained is whether particular servers are available and can be connected to. (The status information will be part of a potential client's Bind.repl result block.)

4.5 Common Server Modules

In a Client/Server architecture, servers operate on a "receive any" basis (cf. Gentleman [9]). The Client/Server architecture as proposed in [18] suggests that all messages (e.g. bind requests and operation requests) received by a server must be handled through a Common Server Module, or CSM. The CSM authorizes bind requests, maintains sessions, "interprets" and dispatches operation requests to the server, and sends replies to clients. The CSM is the server-version of the CCM, and combines the functionality of a session handler and a request handler. We will not digress into the design of the CSM, but some further issues related to it will be mentioned in the later discussion of binding and unbinding. A CSM will be designed in a "server-independent" manner and (thus) needs attribute information of the particular server it supports in a standard format, in order that it can perform the necessary validations of operation requests and replies.

4.6 Maintaining connections

Once a client(A) has succeeded in "binding" a server process (B), the server is prepared to receive and process any suitable operation request which the client sends. Operation request messages are sent under control of the client's CCM, by executing Oper.req commands. A Client/Server architecture should support both blocking ("synchronous") and non-blocking ("asynchronous") requests, and with suitable precautions both types of requests must be allowed over the duration of a single session. It follows that the type (class) of a request must be specified in every operation request block and operation request message, to ensure the proper handling by the Oper.req primitive. It may be necessary to know in a session how many operation request can be outstanding at a time, whether it is blocking or non-blocking. Note that the operation result block of an outstanding operation request contains the necessary status information which the client can check for this purpose (in the model as described here). A Client/Server architecture may allow that a client revokes a (non-blocking) operation request (cf. the Cancel.req primitive in the model described here). In view of the earlier requirements only the unique session-id needs to be supplied as a parameter, as it uniquely identifies the operation request in it that is currently outstanding (if there is one). Of course one needs to specify the effect of a Cancel.req at the server's end very precisely, but quite apparently it can only "intercept" (and cancel) an operation request before its processing has begun and cannot trigger an
“undo” if the processing of the request has started or is completed. Blocking requests cannot be revoked.

In general a server will have to know whether an operation request is blocking or non-blocking at the client’s end, for the purpose of connection and resource management. In some architectures this is not necessary, and servers can be oblivious to the class information of an operation. In this case the server primitives can remain of a simple kind, and indeed need to be no more complicated than “receive” and “send”. Of course servers can perform non-specific actions (like aborting all sessions) and inform clients, with implicit effects for outstanding operation requests. The handling at the client’s end must be part of the .repl primitives. The Client/Server architecture as proposed in [18] suggests that all this is handled by the CCM and CSM modules, thus providing the client with a request/reply mechanism of similar transparency as the well-known RPC.

4.7 Failure semantics

Again the possibility of failures complicates the Client/Server model and (hence) the design of the common interface modules in the Client/Server architecture as proposed in [18]. For example, it must be carefully specified what performance of remote operation requests is required under different conditions of communication or processor failure. If only communication failures can occur, one typically requires either an “at-least-once” semantics or an “exactly-once” semantics for remote operations. (The “exactly-once” semantics is usually preferred and guaranteed in RPC-based remote operations architectures, cf. Birrell & Nelson [4].) If processor failures can occur as well, several other options have been proposed (cf. the detailed taxonomy in Spector [28]). A discussion of reliability issues in the Client/Server model is deferred to a later section. In most Client/Server architectures, all communication requests are handled by a reliable transport level. In this case communication failures (errors and network problems) reported to a client are all connection-oriented.

4.8 Further support services

A Client/Server architecture will usually require that the CCM and CSM provide various additional support services, e.g. various network security functions and resource allocation services. In this paper we will not discuss these services.

5 Remote Function Modules

The Client/Server model is based on a strict separation between client and server functionality. More precisely, the Client/Server model captures all aspects of distributed systems in which processes communicate on a request/reply basis. It follows that the Client/Server model is not restricted to systems with separate client and server nodes and equally well permits that client and server functions reside on a single processor or in a single process. Indeed a Client/Server architecture should permit client-to-client and server-to-server communication using the client-server interaction primitives, and even server-initiated sessions between servers and clients are possible. Thus it is necessary to distinguish between clients and servers as they exist on the network and clients and servers as they exist at some (logical) level of process interaction.
5.1 Building blocks

The apparent lack of uniformity in client and server structures makes it difficult to distinguish common building blocks on which the Client/Server model can be based. At an abstract level all notions related to the theory of concurrent processes (see e.g. Peterson & Silberschatz [26]) can be brought to bear on the Client/Server model, but this clearly does not fully capture all requirements in a concrete network context. In this section we will develop the notion of a Remote Function Module (RFM) and propose it as a basis of the Client/Server model. RFMs are intended primarily for structuring purposes in the design of Client/Server architectures. Thus, designs and specifications should be written in terms of RFMs, but the implementation language need not support the concept as such (as long as its semantics can be realized somehow). Every application, whether client or server (in the network sense), will be composed of a number of local processes and RFMs. At the same time, the necessary support environment for RFMs as we will describe it implies a possible structuring of the CCM and CSM module-concepts.

5.2 Structure of RFMs

A RFM essentially is a set of functions that can be used by or use functions of RFMs at other sites. Thus it consists of a set of entry functions (for local or remote use), a list of remote function names, a set of auxiliary procedures (for internal use), an initialization and driver routine (executed and initialized upon creation of the module) and local data (accessible through the functions and procedures of the module only). RFMs resemble tasks in Ada (see e.g. Barnes [1]) and modules in Modula-2 (see e.g. Wirth [34]), although the implementation model will be different. RFMs are entire self-contained, except for the possible references to remote functions. The use of (logical) function names rather than RFM or server names and addresses gives RFMs a high degree of configuration independence. Also, RFMs can easily be tested in isolation by providing “stubs” for the remote functions it employs. Remote functions can be called by sending off operation request blocks and waiting or checking for a reply, as explained in the previous section. We will discuss momentarily how this should be specified at the RFM level. Note that RFMs, even when they reside on the same processor, have no shared storage and communicate using a mechanism very similar to RPCs.

5.3 Sharing

As its entry functions can be requested ("called") by other RFMs, all RFMs are inherently "shared objects" and thus must provide for shared usage. In typical Client/Server architectures clients are not shared but servers are and, in our terminology, sharing is implemented either by scheduling connections one-at-a-time (single-threaded servers) or by providing a separate instance of the RFM for each connection that requests its services (replicated servers). In the latter case, the separate instances may have to cooperate on the (shared) data of the master RFM. In general the sharing of a RFM requires that an "administrator" or "monitor" is added to handle the simultaneous access by other RFMs (cf. Gentleman [9]). The administrator of a RFM basically maintains the request/reply queues as shared buffers between the RFM and the requesting parties (as in the producer/consumer problem, cf. [25]), and allows the RFM to handle one request at a time. The administrator can implement any desired policy for selecting requests. From now on
we will assume that the shared usage of RFMs is arranged for by standard methods. For a further discussion of the administrator concept, see Gentleman [9]. (In Svobodova et al. [29] the related notion of a "guardian" is introduced for the very same purpose.)

5.4 Connectors

A RFM can only communicate with another RFM if an explicit "connection" (session, association) between the two has been set up. Each connection is to be managed by a new type of (shared) object, a "connector". Basically a connector is an object that governs the message traffic between the two RFMs that it connects and deals with all problems caused by the two RFMs in this respect. To apply the concept in the Client/Server model, it is necessary that we view the "network" as one RFM that is local to every processor and that provides basic transport level services. It is thus no restriction to require that a connector can only connect two RFMs that reside on the same processor. A connector essentially maintains and manages a session between two RFMs in the network sense. The complexity of a connector will depend on the type of RFMs it connects. (The usual connection will be between a RFM and the network, viewed as a RFM.) Yet the main idea of a connector is that it connects local RFMs and thus provides for a high degree of configuration independence and ease of testability. As a RFM, the network is assumed to provide all remote functions by their name and server address. Without much difficulty a connector can maintain a stream of request/reply messages in the case of asynchronous requests and a restriction to one outstanding request at-a-time per connection is not necessary. (The restriction may still be enforced in case of memory limitations on a supporting processor.)

5.5 Configurations

All connection and disconnection requests of the RFMs on a processor are handled by one general (shared) object on this processor, called a "configuration manager" (following the CONIC design of Kramer et al., cf. Sloman & Kramer [27]). We allow that connection requests specify a sub-set of the functions to which access is desired on a RFM. The configuration manager authorizes, administers and dispatches connectors and provides dictionary services like the mapping of function to server names and of server names to network addresses. In the terms of the previous section, the configuration manager is designed to handle all bind and unbind requests. Note that if a RFM wants to communicate with a "client" or a "server", it has to request a connection to the network (as a RFM) for the desired set of client or server functions. The configuration manager will interact with the administrators of both RFMs before dispatching a connector. (In the case of the network RFM this will initialize a network session with the desired client of server.) In the same way connection requests by the network (as a RFM) are handled. Thus all RFMs interface with the configuration manager in an identical manner. The configuration manager can very well be extended and put in charge of more complex actions like the creation (and eventually, the destruction) of copies of a RFM in case of shared access through the "replication" policy. It seems better to handle this at the level of the RFM administrator however. Of course the configuration manager will be in charge of the creation (initialization) and destruction (termination) of RFMs in general.
5.6 Use of RFMs

The few abstractions on which the given view of the Client/Server model is based, are simple and yet sufficiently powerful to accommodate an integral presentation of any Client/Server architecture. For example, authorization requests are easily understood in terms of the request/reply interaction between two suitably defined RFMs. The view requires that all clients and servers are explicitly specified as RFMs, i.e., in terms of the entry functions they provide for local and remote use. All additional services like logging and security services can be fit into the framework. Note that such services can be realized by either enhancing the internal specification of the RFMs (without changing their interface to the other RFMs) or by introducing "filters" between RFMs. Even the functionality of connectors can be enhanced to provide for additional facilities. For example, connectors may be designed for governing one-to-many rather than one-to-one message exchanges and the necessary concurrency control. It should be an interesting research project to develop the complete Client/Server model along the lines suggested. All mechanisms described can be implemented as a "common kernel" and thus are the basis of a possible design of the CCM and CSM modules as proposed in the Client/Server architecture of [18].

5.7 Example

We end this section with a simple example of a possible connector, based on the connection management protocol of Fletcher & Watson [8]. It is assumed that the configuration manager has opened a connection between a client RFM(A) and the network RFM or, indirectly, a server(B) at the transport level. During the connection, A occasionally wants to engage in a session with B and send it a sequence of asynchronous requests. For simplicity we assume that A only expects acknowledgements as replies from B. The connector module is a shared object between A and B that concerns itself only with the management of the message traffic between A and B in the current connection. We only describe the connector at the client's end. Thus, A can append new messages to a shared queue M[Low], ..., M[High-1] in the connector and the connector occasionally reports back which messages have been ack'ed by B. In order to guard for message delays and failures in the network, the connector employs a timer-based regime. It is assumed that the network maintains a "lifetime" for each message and that the network automatically discards a message when its life-time has reached a certain maximum value MPL. (The value of MPL will be chosen such that each message can reach its destination within MPI time-units under normal network conditions.) The connector maintains a timer $St$, which will be reset to $S$ at every "send". A session begins with a first send, and ends the moment that $St \leq o$. With each message $M[i]$ in the shared queue is also associated a so-called retry-timer $Ut[i]$, which is initialized to a certain value $U$ the moment the message is appended by the client and never reset. Intuitively, $U$ bounds the length interval during which a message may be retransmitted (if there is reason to believe that it was not properly received the first time around). In our presentation of the protocol there is no limit on the number of retransmissions allowed for a message during its retry interval. We note that at the receiver's end the connector will likewise maintain a timer $Rt$, which is reset to $R$ at every "receive". A session begins with a first receive (provided it concerns a message with its "start-of-sequence" bit on) and ends, as far as the receiver is concerned, the moment that $Rt \leq o$. It is assumed that the connector sends an ack with $R$ time after receiving a message. From the theory of the Fletcher-Watson protocol it follows that we must require
that $R \geq U + MPL$ and $S \geq 2MPL + R$. The connector at the sender's end provides the following functions

```plaintext
client function send(M);
begin
    Ut[High] := U;
    M[High] := M;
    High := High +1;
end;

server function ack(ESN);
begin
    if ESN > Low then mark index ESN-1 as ack'ed;
    Low := max(Low, ESN)
end;

connector function send;
begin
    if High < Low then skip
    else
        if there exists no i with Low < i < High & Ut[i] > o then skip
        else
            begin
                i := the smallest (or any) index with Low < i < High and Ut[i] > o;
                transmit < (i ≡ Low), i, M[i] > to B;
                St := S
            end
end;

connector function report;
begin
    if Low-1 is marked as ack'ed then
    begin
        transmit < ok, Low-1 > to A;
        mark Low-1 as reported (and unmark it as ack'ed)
    end;
    if Ut[Low] < - 2 MPL - R then
    begin
        transmit < probably lost, Low > to A;
        Low := Low +1
    end
end;
```

Note that the client and network RFMs and the connector driver process compete for functions from this set. Only one function can be executed at-a-time. The notation suggests which functions are accessible to which party. We omit the specification of the connector at the receiver's end (see Fletcher & Watson [8] or Tel [32]). It can be shown that the combined connectors guarantee that there is no undetected loss of messages and
no problem with duplicates. Note that requests that are reported as "possibly lost" may actually have arrived in B. But every request that is lost will be reported to A (and leads to a new start of the message sequence). The precise parameter values and the connector's regime of function calls determine to large extent the efficiency of the protocol. The same type of protocol can be recognised in the Client/Server architecture proposed in Janson et al. [15]. In the Amoeba system (Mullender [24]) a version was used in the design of a transaction protocol for the Client/Server model, with an even more elaborate use of timers and the "no reply" policy used in the example above replaced by a strict request/reply interchange per transaction.

6 Binding and Unbinding

We continue our exploration of the Client/Server model and discuss the issues dealing with binding and unbinding. (We will use the Client/Server model as understood up to section 4 and make no reference to the abstractions proposed in section 5.) In the Client/Server model, binding is the act of connecting a "logical reference" to a "physical object". Binding involves naming and server location, port determination and activation, and connection opening. Binding leads to a connection between client and server, and facilitates the exchange of request/reply messages between the two. Unbinding is the act of disconnecting (releasing) a binding.

6.1 Naming

The naming problem in distributed systems is well-known. Names are needed to (uniquely) identify objects and operations but are often required to be location-independent. It follows that a mechanism (a mapping) is required to map names to addresses, in some way or another. The naming policy adopted in a Client/Server model is important, and must be convenient at the application level and efficient when it comes to binding. (Also, the uniqueness or non-ambiguity of a name should be easy to verify during name assignment.) Mullender [24] distinguished four different naming policies that prevail in distributed systems for determining system names for services:

(i) domain naming: a system name consists of a host name and an object name on this host. The policy is simple but not very flexible, as it does not allow objects to move through the network without changing their system name.

(ii) global naming: a system name is any globally unique name. In this policy names are not bound to the location of an object, but now a (centralized or distributed) name server is needed to complete the binding. The name server should be at a well-known address, and is in charge of maintaining the name-location information. The binding procedure may augment a system name by a port name (both at the client's and the server's end) to further identify a temporary access point for the connection.

(iii) static port naming: a system name is a "well-known" port. In this policy ports are associated with services rather than objects. A client wishing to connect to a service only needs to know the (network-wide) well-known port. The service provider sees to it that an object (a process) is maintained for every port that is accessed.
(iv) dynamic port naming: a system name is a port that is created and managed by a
"user". When a server decides to offer a service to the network, it generates a port for
that service, announces the port’s name to potential clients, and starts “listening”
to the port (i.e., it is willing to engage in a connection at this port). The action
of the server can be triggered by a signal from a particular client, which may have
allocated a reply port at its end (but which the server does not have to know).

6.2 Binding

Binding normally refers to the mapping of user-level names to system names. A system
name should be sufficient for the transport level to determine host addresses and routes.
Distributed name servers and dynamic port naming are current research topics. Port
determination policies for the Client/Server model are described in more detail in e.g.
Bershad et al. [3]. We will assume that the basic client and server processes are “pre-
activated”, and do not go into the details of “activation through binding” (or “auto-
activation”). In dynamic (extendible) networks the naming problem is more complicated,
because new hosts must be assigned new and preferably compact names that are unique on
the network. Typical solutions either use centralized name servers or employ a distributed
scheme in which existing nodes have sets of names available that can be assigned to new
nodes that are attached to it. (The latter leads to “dynamic domain naming”.)

6.3 Establishing connections

The next, essential part of binding concerns the opening of a connection (session, associ-
ation) between the client and the server. Naming and port determination are implicit in
the actions taken to establish a connection. The common protocols for binding (see e.g.
Knowles et al. [16]) are all based on a 2-way handshake, using the following primitives:

(i) CONNECT.request

(ii) CONNECT.indication

(iii) CONNECT.response

(iv) CONNECT.confirm

(Only in datagram networks should a 3-way handshake be used, for reliably opening and
closing connections. We do not consider this in the present paper.) The primitives are usu-
ally prefixed with a letter that identifies the layer of service in which they are implemented
(e.g. T for “transport layer”). Thus, at the session level a connection can be opened by per-
forming a T.CONNECT.request and waiting until a corresponding T.CONNECT.confirm
is performed. The ISO standard (see e.g. [14]) specifies in detail what components can be
distinguished in the whole process. The T.CONNECT.request can be viewed as a primitive
but will be implemented using lower level CONNECT.request and CONNECT.confirm
primitives. Using the model from before, the T.CONNECT.request (at the client’s end)
must begin by naming a reply port and setting up a connection record. Now two policies
can be followed, depending on the model assumptions:

(i) connect-to-server: the client possesses the well-known address of the server. It sends
the proper TPDU ("Transport-Protocol-Data-UNIT") to the server and triggers its
T.CONNECT.indication procedure.
(ii) connect-to-port: the client waits until the service it wants announces its port name on the network. When the port is known, the client sends off its T.CONNECT.request to the port.

In both cases the T.CONNECT.indication primitive at the server will name a request port which the potential client should address when sending messages. In general the T.CONNECT.request primitive is used to convey a client’s requirements (in terms of required resources), and the other primitives arrange for these requirements and inform the requester of the quality of service that will be provided (if at all).

Once a connection (session, association) has been established, “data” can be transferred. In most Client/Server architectures, data can be transferred in the standard “2-way simultaneous” mode. No tokens are needed and, indeed, no token management primitives need to be provided. Logging services are required for recovery purposes, and “minor/major synchronization” must be provided for in the session maintenance protocol.

6.4 Releasing connections

The protocols for unbinding (release, close) are now based on a slightly different handshaking principle, using the following primitives:

(i) DISCONNECT.request

(ii) DISCONNECT.indication

Either party in a connection can initiate a DISCONNECT.request, but normally it will be the connection initiator that does so after its use of the desired services. A DISCONNECT.request by the client results in an ACK as soon as the server detects that there are no outstanding messages of the client. The server subsequently issues a DISCONNECT.request from its end and a similar procedure repeats. Ultimately both parties close the connection.

6.5 Unbinding

A Client/Server architecture must likewise provide primitives for unbinding, following the ISO standards. Unbinding is achieved by releasing the connection record and removing the port that existed for the connection, by suitable actions at the sending and the receiving end.

7 Managing Atomic Actions

The Client/Server model assumes that clients and servers interact strictly on a request/reply basis. In applications it may be desirable that a client and a server interact in a more complex manner during a session and engage in an activity (a set of operations or “an action”) that affects the information stored at the client and the server simultaneously and in an indivisible manner. Activities of this kind are called “transactions” or “atomic actions”. Atomic actions would pose no particular problem if it weren’t for the fact that in all realistic applications “exceptions” and “failures” (like link failures or site crashes) can occur during an atomic action. We will use the term “failure” to refer to any abnormal condition that arises during an atomic action. The possibility of failures requires that
the effects of an atomic action must be recoverable at all times throughout its elaboration, until the atomic action can be regarded as “safely completed” at both ends. The implementation of atomic actions thus requires two basic facilities (see e.g. Gray [10]):

(i) a recovery mechanism, i.e., a mechanism for “undo-ing” the effect of one or more atomic actions. Recovery mechanisms are always based on the use of logs that record information on the processing of atomic actions at each site, and on a method for effectuating a rollback. Logs must be recoverable in case of failures and thus must be kept on “stable storage”.

(ii) an atomic commit protocol, i.e., a protocol for detecting “safe completion” and committing the effects of an atomic action at both ends. Atomic commit protocols are atomic actions and thus must be recoverable themselves.

The occurrence of a failure at some site does not necessarily imply that the atomic actions in progress must all be aborted and rolled back. It may be possible for a node to recover to a consistent state, based on information in its log. (This is called “independent recovery”.) After recovery, a site may wish to have an atomic action re-started.

7.1 Implementation of atomic actions

The implementation of atomic actions in the context of possible failures is a well-studied problem. In this section we will only discuss some aspects of the atomic commit problem, as it will appear in enhanced Client/Server architectures. An excellent introduction to concurrency control and recovery in distributed databases was given by Bernstein et al. [2]. Also, a detailed recommendation for the implementation of atomic commit protocols appears in the CCR standard of ISO [13]. It suggests that atomic commit protocols follow some version of the well-known 2-phase commit protocol due to Gray [10] and Lampson & Sturgis [19], and use the following primitives:

(i) C - BEGIN
(ii) C - PREPARE
(iii) C - READY
(iv) C - REFUSE
(v) C - COMMIT
(vi) C - ROLLBACK
(vii) C - RESTART

Our main goal will be to give a more refined and complete presentation of the Individual Commit protocol than is usually given (cf. [2,6]. As it will require no extra effort, we will describe the protocol for the more general case of a client-initiated atomic action that involves multiple servers. It is assumed that the client remains the “coordinator” (or “superior”) of the atomic action and thus of the atomic commit protocol. The client will only initiate the commit protocol (with a C-PREPARE primitive) if it has reason to do so, i.e., if the activity of the atomic action at its own site has ended (which implies
that the servers have provided their operation results insofar as needed by the client). At all times during the atomic action, the client and the servers must be ready to honor a C-ROLLBACK or C-RESTART request from any party in the atomic action. Note that the servers only communicate with the client, but not with each other (during the atomic actions).

7.2 Atomic commit protocols

Applied to an atomic action, an atomic commit protocol is a distributed algorithm for the client and the servers that should guarantee that they all commit or all abort the atomic action. Following Bernstein et al. [2] the situation for an atomic commit protocol can be rephrased in more precise terms as follows. Each party (client or server) may cast exactly one of two votes: C-READY ("yes") or C-REFUSE ("no"), and can reach exactly one of two decisions: C-COMMIT ("commit") or C-ROLLBACK ("abort"). An atomic commit protocol must satisfy the following requirements:

AC1 : A party cannot change its vote after it has cast a vote.
AC2 : All parties that reach a decision reach the same decision.
AC3 : A party cannot change its decision after it has reached one.
AC4 : A C-COMMIT decision can be reached only if all parties voted and voted C-READY.
AC5 : If there are no failures and all parties voted C-READY, then the decision C-COMMIT will be reached by all parties.
AC6 : Consider any execution of the protocol in the context of permissible failures. At any point in this execution, if all current failures have been repaired and no new failures occur for a sufficiently long period of time, then all parties will eventually reach a decision.

The requirements can be viewed as the minimal correctness criteria for atomic commit protocols. (Except for AC1, the requirements are taken from Bernstein et al. [2].) AC1 through AC5 can usually be satisfied quite easily, but AC6 requires a suitable recovery procedure to be part of the protocol. Note that after voting C-READY, a party (client or server) can not be certain of what the decision will be until it has received sufficient information to decide. Until that moment, we say that the party is "uncertain". If a failure occurs that cuts an uncertain party off, then this party is said to be blocked. A blocked party cannot reach a decision until after the connection to the other parties has been restored. Blocking is usually concluded if no messages arrive during a certain timeout interval within the uncertain period. (Heuristic commit protocols allow a blocked party to make a calculated guess of the decision. In the Individual Commit protocol proposed in [18] a blocked client is allowed to commit.) A different situation arises when a party fails (crashes) during its uncertainty period. In this case a more involved recovery procedure may have to be followed (see Bernstein et al. [2] or below).

7.3 The (2-phase) individual commit protocol

The standard (2-phase) atomic commit protocol is as follows, in terms of the recommended CCR primitives. (Note that the desired steps of the recovery procedure after failure are
part of the overall protocol, but these are not included in the basic specification below.)

Individual Commit Protocol

Client’s Commit

Phase 1

  c-0. Vote ready;
  
  c-1. Write “prepare” record to the Log;
  
  c-2. Send C-PREPARE messages to all servers; activate timer;
  
  c-3. {Await answer messages (C-READY or C-REFUSE) from all servers using
        a timer and act as follows}

        Case condition of
        c-3.1. Timeout or C-REFUSE message received:

            begin
            Write “rollback” record to the Log;
            Send C-ROLLBACK messages to all servers;
            C-ROLLBACK
            end;

        c-3.2. All servers answered and answered C-READY:
            continue with Phase 2

      end;
      {End of Phase 1}

Phase 2

  c-4. Write “commit” record to the Log;
  
  c-5. Send C-COMMIT messages to all servers; activate timer;
  
  c-6. {Await answer messages (ACK) from all servers using a timer and act
        as follows}

      CASE condition of
      c-6.1. Timeout: Write “incomplete” record to the Log;
      c-6.2. All servers answered (ACK):

        Write “complete” record to the Log;

      end;

  c-7. C-COMMIT;
  {End of Phase 2}

Server’s Commit

Phase 1

  s-1. Await a C-PREPARE message from the client using timer;

      {The server will only pass this point if it has indeed received a
       C-PREPARE message from the client or timed out}

  s-2. If not timed out then Vote ready or refuse;

  s-3. Case condition of

      s-3.1. ready:

          begin
          Write “ready” record to the Log;
          Send a C-READY message to the client;

      end;
continue with Phase 2

end;

s-3.2. refuse:

begin
  Write "refuse" record to the Log;
  Send a C-REFUSE message to the client
end

s-3.3. Timeout:

  Write "refuse" record to the Log

end;

{End of Phase 1}

Phase 2

s-4. Await a decision message (C-COMMIT or C-ROLLBACK) from the client using timer;

{The server will only pass this point if it has indeed received a decision message from the client or times out}

s-5. Case condition of

s-5.1. a C-COMMIT message was received:

begin
  Write "commit" record to the Log;
  Send ACK message to the client;
  C-COMMIT
end;

s-5.2. a C-ROLLBACK message was received:

begin
  Write "rollback" message to the Log;
  C-ROLLBACK
end;

s-5.3. Timeout:

  take whatever action to deal with blocking

end;

{End of Phase 2}

Although it is not explicitly specified, each party can unilaterally decide for a C-ROLLBACK at any time if it has not (yet) voted "ready". We assume that the Client’s Commit protocol is only initiated by the client after it has voted "ready". If a client never votes or votes "refuse", then it never sends a C-PREPARE message and the servers automatically time out eventually in their protocol. (We could have treated the client as a server in the protocol, but have chosen not to do so in order to save messages.) We require that C-COMMIT messages are ack’ed, but this can be omitted from the protocol without any harm. (We do not require that C-ROLLBACK messages are ack’ed, but could incorporate it easily if desired.) Voting essentially amounts to determining whether the local activities in an atomic action have ended and properly been logged or not, but we only need it as an "abstract" operation. Where ever a "Log" is specified in the protocol, the local (client or server) log is meant. The Individual Commit protocol as desired reduces to the protocol outlined in [18] for the case of one server. A “Phase 2” is not entered automatically after
a "Phase 1", but only on an explicit "continue".

7.4 Correctness proof

It should be an interesting research topic to develop a completely formal "correctness proof" for the Individual Commit protocol. We outline a less formal proof here. We refer to line-numbers as c-0, c-1, etcetera.

Theorem 7.4.1 If no timeouts and no failures occur, then the Individual Commit protocol is correct.

Proof.
The requirements AC1 through AC5 are trivially satisfied. For AC2, note that if any party (client or server) decides spontaneously for C-ROLLBACK when it can, then all parties must follow suit eventually and cannot decide for anything else. AC6 is vacuously true. □

The next step is to consider the possibility of timeouts and a limited type of failures, namely "loss of protocol messages". In all cases except one, message loss necessarily leads to a timeout and thus it is sufficient to consider the latter only. The one exceptional case can arise in c-3, when some messages (including a C-REFUSE) have arrived but some have not and the clients acts on the C-REFUSE. In this case the client acts just like it would have in the case of timeout (line c-3.1.). Note that heavily delayed messages are considered "lost".

Theorem 7.4.2 If timeouts and loss of messages can occur but no server gets blocked, then the Individual Commit protocol is correct.

Proof.
The requirements AC1, AC3 and AC4 are trivially satisfied. For AC2, observe the following. If a server times out on s-1, it will never send a message and can only decide C-ROLLBACK (if it ever decides, cf. s-3.3.). The client necessarily executes c-3.1. and decides for C-ROLLBACK too. Other servers either time out on s-1 as well or receive the C-ROLLBACK decision in s-4. By assumption no server times out on s-4 (the blocked case). If the client times out on c-3.1., then it decides C-ROLLBACK and the servers can only reach the same decision by the very same argument. If the client times out on c-6.1., the only possible decision in the system is C-COMMIT and all servers must be in Phase 2. As we assume no blocking, all servers will eventually decide C-COMMIT. Thus AC2 is satisfied and, by the latter argument, AC5 as well. AC6 is vacuously true. □

In order to get any further we must some how deal with the problem of blocking. A blocked server timed out on s-5.3. and thus knows that it is blocked, but it is uncertain of the decision that may have been reached. In order to keep the Individual Commit protocol correct, a Server's Commit Termination Protocol must be added. The purpose of the Server's Commit Termination protocol is to enable a blocked server to determine the (apparent) decision reached in the system. Several possible strategies for a successful Server's Commit Termination protocol have been proposed, all based on polling. (For example, if another server can be reached and it appears to have timed out on s-1, then the blocked server can decide C-ROLLBACK.) But no Server's Commit Termination protocol can guarantee that it will remove the possibility of blocking. (For example, if a blocked
node is cut off permanently, it will forever remain uncertain.) We assume that any server can eventually reach the client again and thus a simple polling of the client will do as a Server's Commit Termination protocol (cf. requirement AC6). We conclude the following result.

**Theorem 7.4.3** If timeouts and loss of messages can occur, then the Individual Commit protocol enhanced with the Server's Commit Termination protocol is correct.

The final step is to allow timeouts and arbitrary failures, i.e., "loss of protocol messages" and "site crashes". If a site crashes permanently, the Individual Commit protocol will simply continue in the remaining sites and perform as if the messages of the crashed site are lost from some point onwards. The Server's Commit Termination protocol should be extended in this case and somehow detect permanent crashes of the client. In general there is no guaranteed solution that avoids blocking, if permanent crashes can occur. Thus we assume that each site that crashes eventually recovers and resumes the commit protocol. We also assume that a site can actually recover to the point where it crashed, using the information in its log. The only problem now is to determine how to continue with the commit protocol, knowing that the other sites may have advanced in it after the crash. We present a possible recovery protocol below.

**Commit Recovery Protocol**

**Client's Commit Recovery**

1. If the client crashed before c-4 then
   
   begin
   Write "rollback" record to the Log;
   Send C-ROLLBACK messages to all servers;
   end;

2. If the client crashed after c-4 but before c-6.2. then
   
   begin
   Write "commit" record to the Log;
   Send C-COMMIT messages to all servers; activate timer;
   {Await answer messages from all servers using timer and act as follows}
   Case condition of
   Timeout: Write "incomplete" record to the Log;
   All servers answered: Write "complete" record to the Log
   end;
   C-COMMIT
   end;

3. If the client crashed after c-6.2. then
   
   begin
   perform c-7 if necessary
   end;

**Server's Commit Recovery**

1. If the server crashed before s-4 and without having executed s-3.1.
then
begin
C-ROLLBACK
end;
sr-2. If the server crashed after s-4 while being uncertain
then
begin
use the Server's Commit Termination protocol to remove blocking
end;
sr-3. If the server crashed after s-4 while being certain
then
begin
perform remaining activity if necessary in the C-COMMIT or
C-ROLLBACK
(whatever applies)
end;

We assume that Write/Send commands in the Individual Commit protocol are atomic, and thus no crash occurs "in between" a Write and the subsequent Send. (While the assumption is reasonable, it would nevertheless be of interest to analyze the protocol if this assumption is not made.)

**Theorem 7.4.4** The Individual Commit Protocol enhanced with the Server's Commit Termination protocol and the Commit Recovery Protocol is a correct atomic commit protocol.

**Proof.**
If the client crashes before c-4, then each server either has C-ROLLBACK as the only option is uncertain. Thus cr-1 is a correct recovery action, in the sense of satisfying the requirements. If the client crashes in its Phase 2, then each server has either decided C-COMMIT or is uncertain. Replaying Phase 2 (in so far as it its Phase 1 and without sending a C-READY message, then the remaining sites will have progressed on the assumption that its messages are C-REFUSE or "lost". It means that the remaining sites are on their way to decide C-ROLLBACK, and sr-1 is fully consistent with this. If a server crashed after having sent a C-READY message, then either it had decided (and the decision is recovered) or is uncertain. Thus sr-2 and sr-3 are the actions to take. With the earlier analyses it easily follows that the complete protocol satisfies AC1 through AC6 and hence is correct. □
References


