

APPLICATIONS FOR GROUP MANAGEMENT TO MANAGE CPS STABILITY

by

STEPHEN JACKSON

A DISSERTATION

Presented to the Faculty of the Graduate School of the

MISSOURI UNIVERSITY OF SCIENCE AND TECHNOLOGY

In Partial Fulfillment of the Requirements for the Degree

DOCTOR OF PHILOSOPHY

in

COMPUTER SCIENCE

2014

Approved by

Dr. Bruce McMillin, Advisor

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ABSTRACT

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Abstract

Cyber-physical systems (CPS) are an attractive option for future development of critical infrastructure systems. By supplementing the traditional physical network with cyber control, the performance and reliability of the system can be increased. In these networks, distributing the cyber control offers increased redundancy and availability during fault conditions. However, there are very few works which study the effects of cyber faults on a distributed cyber-physical system. These are of a particular interest in the smart grid environment where outages and failures are very costly. By examining the behavior of a distributed system under fault scenarios, the overall robustness of the system can be improved by planning characteristics and responses to faults that allow the system to continue operating in difficult circumstances.

This work presents a technique for modelling the availability of a cyberphysical system. This is of high value because there is a limited amount of research that describes that models the availability of distributed systems, and CPSs in general. This approach encodes the behavior of a dynamic reconfiguration system of a CPS as a continuous-time Markov chain. These models provide an estimate of when a cyberphysical system will be able to do work and when it is reconfiguring. Generated models are presented and evaluated against data collected from actual runs of the system. Using these models, we will be able to develop methods and guarantees to quantify and protect operation of a CPS.

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TABLE OF CONTENTS

	Page
ABSTRACT	iii
ACKNOWLEDGMENTS	i
LIST OF ILLUSTRATIONS	v
LIST OF TABLES	vi
 SECTION	
1 INTRODUCTION	1
2 RELATED WORK	3
2.1 Failure Classification	3
2.1.1 Crash or Fail-Stop Failure	3
2.1.2 Omission Failure	3
2.2 Virtual Synchrony	4
2.2.1 Process Groups	5
2.3 Extended Virtual Synchrony	7
2.3.1 Comparison To DGI	9
2.4 Isis (1989) and Horus (1996)	10
2.4.1 Group Model	10
2.5 Transis (1992)	11
2.6 Totem (1996)	12
2.7 Spread	12
2.8 Failure Detectors	13

3	BACKGROUND THEORY	15
3.1	FREEDM DGI	15
3.2	Broker Architecture	16
3.2.1	Sequenced Reliable Connection	17
3.2.2	Sequenced Unreliable Connection	17
3.2.3	Real Time	18
3.3	Group Management Algorithm	20
3.4	Network Simulation	21
3.5	Markov Models	22
3.5.1	Continuous Time Markov Chains	22
3.5.2	Constructing The Markov Chain	24
3.5.3	Assumptions	30
4	RESULTS	31
4.1	Initial Results	31
4.1.1	Sequenced Reliable Connection	31
4.1.2	Sequenced Unreliable Connection	34
4.2	Markov Models	35
4.2.1	Initial Model Calibration	35
5	CONCLUSION	38
5.1	Selecting A Schedule	38
5.2	Correctness of an Installed Configuration	42
5.3	Accuracy and Scope of The Model	42
5.4	Deliverables	44

APPENDICES

BIBLIOGRAPHY	45
GLOSSARY	48

LIST OF ILLUSTRATIONS

Figure	Page
3.1 Real Time Scheduler	19
3.2 A diagram showing a partial Markov chain for failure detection	26
3.3 A diagram showing a partial Markov chain for an election	28
4.1 Time in group over a 10 minute run for two node system with 100ms resend time	32
4.2 Time in group over a 10 minute run for two node system with 200ms resend time	32
4.3 Average size of formed groups for the transient partition case with 100ms resend time	32
4.4 Time in group over a 10 minute run for the transient partition case with 100ms resend time	32
4.5 Average size of formed groups for the transient partition case with 200ms resend time	33
4.6 Time in group over a 10 minute run for the transient partition case with 200ms resend time	33
4.7 Time in group over a 10 minute run for two node system with 100ms resend time	34
4.8 Time in group over a 10 minute run for two node system with 200ms resend time	34
4.9 Comparison of in-group time as collected from the experimental plat- form and the simulator (1 tick offset between processes).	35
4.10 Comparison of in-group time as collected from the experimental plat- form and the simulator (2 tick offset between processes).	35
4.11 Comparison of in-group time as collected from the experimental plat- form and the time in group from the equivalent Markov chain (128ms between resends).	37
4.12 Comparison of in-group time as collected from the experimental plat- form and the time in group from the equivalent Markov chain (64ms between resends).	37

LIST OF TABLES

Table	Page
4.1 Error and correlation of experimental data and Markov chain predictions	37
5.1 Comparison of two proposed schedules, A and B	40

1. INTRODUCTION

A robust cyber-physical system should be able to survive and adapt to communication network outages in both the physical and cyber domains. When one of these outages occurs, the physical or cyber components must take corrective action to allow the rest of the system to continue operating normally. Additionally, process may need to react to the state change of some other process. Managing and detecting when other processes have failed is commonly handled by a leader election algorithm and failure detector.

In the smart grid domain, leader elections are an attractive option for autonomously configuring cyber components so processes can coordinate together to manage the physical system. Algorithms such as [1] and [2] are distributed algorithms for managing power in a smart grid that rely on an assumption that a group of processes will be able to coordinate together.

FREEDM (Future Renewable Electric Energy Delivery and Management) is a Smart Grid project focused on the future of the electrical grid. Major proposed features of the FREEDM network include the Solid State Transformer (SST), local energy storage, and local energy generation[3]. This vein of research emphasizes decentralizing the power grid: making it more reliable by distributing energy production resources. Part of this design requires the system to operate in islanded mode, where portions of the distribution network are segmented from each other. However, there is a major shortage of work on the effects cyber outages have on CPSs [4] [5]. However, research that has been done, such as [6] indicate that cyber faults can cause a physical system to apply unstable settings.

This work begins with observations on the effects of network unreliability on the group management module of the Distributed Grid Intelligence (DGI) used by

the FREEDM smart-grid project. This system uses a broker system architecture to coordinate several software modules that form a control system for a smart power grid. These modules include: group management, which handles coordinating processes via leader election; state collection, a module which captures a global system state; and load balancing which uses the captured global state to bring the system to a stable state.

This work presents the initial steps to better understanding and planning for these faults. As part of an analysis of omission failures, we present an approach in modeling the grouping behavior of a system using Markov chains. These chains produce expectations of how long a system can be expected to stay in a particular state, or how much time it will be able to spend coordinating and doing useful work over a period of time. Using these measures, the behavior of the cyber system can be specified to prevent faults.

In Chapter 2, we outline the state of the art: other distributed frameworks and works similar to our own. Chapter 3 provides a background of the architecture of DGI and the properties of the created Markov models. Next, Chapter 4 shows the results that have been collected so far. The final chapter summarizes the work and research goals.

2. RELATED WORK

2.1. FAILURE CLASSIFICATION

Processes can fail in a variety of ways. In this work we focus on how cyber processes can fail. There are a variety of faults that have been classified. In this work, we focus on two different modes of failure:

2.1.1. Crash or Fail-Stop Failure. A crash failure or its more generalized form, a fail-stop failure, describes a failure in which a process stops executing. In general, this is considered to be an irreversible failure, since a process typically does not resume from a crashed state. There are categorizations of crash failures, called napping failure where a process will appear to have crashed for a finite amount of time before resuming normal operation. Crash failures are impossible to detect with absolute certainty in an asynchronus system. Processes can be suspected by other processes through the use of challenge/response messages or heartbeat messages that allow a process to prove that it has not crashed yet. A system that can handle a crash failure is implied to be able to handle a fail-stop failure.[7]

The fail-stop failure has three properties [7]:

1. When a failure occurs the program execution is stopped.
2. A process can detect when another fail-stop process has failed.
3. Volatile storage is lost when the fail-stop process is stopped.

2.1.2. Omission Failure. An omission failure occurs when a message is never recieved by the reciever. Omission failures can occur when the communication medium is unavalable, or when the latency of message exceeds a timeout for it's expected delivery. Protocols like TCP do not tolerate omission failures: a packet is

resent until it is acknowledged by the receiver. If the acknowledgement never comes, the connection is closed. As a contrast, UDP assumes that any datagram could be lost and it is the responsibility of the programmer to handle missing datagrams appropriately. An omission failure can have the same observable effects as a napping failure in some situations. [7]

2.2. VIRTUAL SYNCHRONY

Virtual synchrony (and its original implementation, Isis[8]) is an execution model for distributed systems. Virtual synchrony is based on the idea of synchronous execution where each process acts based on a globally shared clock. Synchronous systems are the easiest to program for, but true synchrony is not feasible in most circumstances. Instead, the virtual synchrony execution model was developed.

Virtual synchrony is a model of communication and execution that allows the system designer to emulate synchronous execution. Although the processes do not execute tasks simultaneously, the execution history of each process cannot be differentiated from a trace where tasks are executed simultaneously.

Consider a distributed system where events can be causally related based on their execution on the local processors and communication between processes, as defined in [8, p .101].

1. Let $e \rightarrow e'$ be a casual relationship between events e and e'
2. If e and e' are events local to a process P_i and e occurs before e' , then $e \rightarrow e'$
3. If $e = \text{send}(m)$ and $e' = \text{deliver}(m)$ for the same message m , then $e \rightarrow e'$

This defines a dependence and causality relation between events in the system. If two events cannot be causally related (that is $e \not\rightarrow e'$ and $e' \not\rightarrow e$ then the events are considered concurrent. From this, one can define an execution history H . A pair

of histories H and H' are equivalent if, the history of a process p , $H|_p$, cannot be differentiated from another history, $H'|_p$, based on the casual relationships between the events in the history. [8, p .103] Additionally, a history is considered complete if all sent messages are delivered and there are no casual holes. A casual hole is a circumstance where an event e is casually related to e' by $e' \rightarrow e$ and e appears in a history, but e' does not. A virtually synchronous system is one where each history in the system is indistinguishable from all histories produced by a synchronous system. [8, p .104]

2.2.1. Process Groups. Virtual synchrony models also support process groups. Although each implementation of a virtually synchronous system applies a different structure to the way processes are grouped, implementations share common features. A process group in a virtual synchronous system is commonly referred to as a view. A view is a collection of processes that are virtually synchronous with each other. A view refers to a specific group: the set of processes in that view, and a leader if there is one. A group is a general descriptor for a collection of processes. Process groups in virtually synchronous systems place obligations on the delivery of messages to members of a view. A history H of a view is legal if [8, p .103]:

1. Given a function $time(e)$ that returns a global time of when the event occurred e , then $e \rightarrow e' \Rightarrow time(e) < time(e')$.
2. $time(e) \neq time(e') \forall e, e' \in H|_p$ (where $e \neq e'$) for each process.
3. A change in view (group membership) occurs at the same logical time for all processes in the view.
4. All multicast message deliveries occur in the current view of a group. That is, if a message is sent in a view, it is delivered in that view, for every process in that view.

A process group interacting via message passing creates a legal history for the virtually synchronous system. However, processes can fail. Virtual synchrony systems have several properties to handle fail-stop failures[8, p .102]:

- The system employs a membership service. This service monitors for failures and reports them to the other processes in the view.
- When a process is identified as failing it is removed from the groups that it belongs to and the remaining processes determine a new view.
- After a process has been identified as failing, no message will be received from it.

Virtual synchrony's failure model uses the fail-stop failure model. Virtual synchrony does not guarantee how messages will be delivered while transitioning between views. Virtual synchrony only guarantees that view changes are totally ordered with respect to the regular messages sent in the view. A message first sent in an old view may be delivered in a new view. If partitions were allowed to join, the ordering of messages still outstanding from the original views is ambiguous.

Regular messages in virtual synchrony, those which do not announce view changes, can be casually ordered or totally ordered. Additionally, virtual synchrony supports two classes of multicast delivery guarantees: uniform and non-uniform. The uniform property obligates that if a multicast message is delivered to a process in a view, it is delivered to all processes that view. Non-uniform multicast is a multicast that does not guarantee the uniform delivery property.

2.3. EXTENDED VIRTUAL SYNCHRONY

One of the major shortcomings of the original virtual synchrony model lays in how it handles network partitions. In virtual synchrony, when the network is partitioned, only processes in the primary partition were allowed to continue. Processes that were not in the primary partition could not rejoin the primary partition without being restarted: the process joining the view must be a new process so that there are no message delivery obligations for that process in the view. [9] presented an improved version of virtual synchrony, dubbed extended virtual synchrony. Extended virtual synchrony is compatible with the original virtual synchrony, and can implement all the functionality and limitations of the original design. Because it supports virtual synchrony, extended virtual synchrony has become the basis for a number of related frameworks that have been designed since Isis including Horus, Totem, Transis, and Spread.

Extended virtual synchrony places obligations on the message delivery service, described informally below [9]:

1. As defined in Virtual Synchrony, events can be causally related. Furthermore, if a message is delivered, the delivery is casually related to the send event for that message.
2. If a message m is sent in some view c by some process p then p cannot send m in some other view c'
3. A view is “installed” when process recognizes a change in group membership.
4. If not all processes install a view or a process leaves a view, a new view is created. Furthermore, views are unique and events occur after a view’s creation and before its destruction. Messages which are delivered before a view’s destruction must be delivered by all processes in that view (unless a process fails). Similarly,

a message delivery which occurs after the creation of a view must occur after the view is installed by every process in the view.

5. Every sent message is delivered by the process that sends it (unless it fails) even if the message is only delivered to that process.
6. If two processes are in sequential views, they deliver the same set of messages in the second view.
7. If the send events of two messages are casually related, if the second of those messages is delivered, the first message is also delivered.
8. Messages delivered in total order must delivered at the same logical time. Additionally, this total order must be consistent with the partial, casual order. When a view changes, a process is not obligated to deliver messages for processes that are not in the same view.
9. If one process in a view delivers a message, all processes in that view deliver the message, unless that process fails. If an event which delivers a message in some view occurs, then the messages that installed that view were also delivered.

It also provides additional restrictions on the sending of messages between two different views (Item 2) and an obligation on the delivery of messages (Item 4). The concept of a primary view still exists within the extended virtual synchrony model and is still described by the notion of being the largest view. Since views can now partition and rejoin the primary partition the rules presented above also allow the history of the views with respect to the regular messages in the primary partition to be totally ordered. Additionally, two consecutive primary views have at least one process that was a member of each view.

To join views, processes are first informed of a failure (or joinable partition) by a membership service. Processes maintain an obligation set of messages they have

acknowledged but not yet delivered. After being informed of the need to change views, the processes begin buffering received messages and transition into a transitional view. In the transitional view messages are sent and delivered by the processes to fulfill the causality and ordering requirements listed above. Once all messages have been transmitted, received and delivered as needed and the process' obligation set is empty, the view transitions from a transitional view to a regular view and execution continues as normal.

2.3.1. Comparison To DGI. The DGI places similar but distinct requirements on execution of processes in its system. The DGI enforces synchronization between processes that obligates each active process to enter each module's phase simultaneously. This is fulfilled by using a clock synchronization algorithm. The virtual synchrony model, at its most basic, does not require a clock to enforce its ordering.

However, this simplifies the DGI fulfilling its real time requirements. Processes must be able to react to changes in the power system within a maximum amount of time. These interactions do not require interaction between all processes within the group. Additionally, this allows for private transactions to occur between DGI processes.

The DGI has also been implemented using message delivery schemes that are unicast instead of multicast, since this is the easiest to achieve in practice. A majority of systems implementing virtual synchrony use a model where local area communication is emphasized, with additional structures in place to transfer information across a WAN to other process groups.

Groups in DGI are not obligated to deliver any subset of the messages to any process in the system. Some of the employed algorithms in DGI need all of the messages to be delivered in a timely manner, but they do not require the same message to be delivered to every process in the current design of the system.

2.4. ISIS (1989) AND HORUS (1996)

A product of Dr. Kenneth Birman and his group at Cornell, Isis [8] and Horus [10] are two distributed frameworks which are comparable to the DGI. Although these projects are no longer actively developed, Isis is the foundation which all other virtual synchrony frameworks are based.

Isis was originally developed to create a reliable distributed framework for creating other applications. As Isis was one of the first of its kind, the burden of maintaining a robust framework that met the development needs of its users eventually led Birman's group to create a newer, updated framework called Horus, which implemented the improved extended virtual synchrony model. However, Isis and Horus largely implement the same concepts.

Both Horus and Isis are described as a group communications system. They provide a messaging architecture which clients use to deliver messages between processes. The frameworks provide a reliable distributed multicast, and a failure detection scheme. Horus offers a modular design with a variety of extensions which can change message characteristics and performance as needed by the project.

2.4.1. Group Model. In Horus and Isis, a group is a collection of processes that can communicate with one another to do work. Multicasts are directed to the view, and are guaranteed to be received by all members or no members. Over time, due to failure, the view will change. Since views are distributed concurrently and asynchronously, each process can have a different view. Horus is designed to have a modular communication structure composed of layers, which allows the communication channel to have different properties that will affect which messages will be delivered in the event of a view change. Horus' layers allow developers to go as far as to not use the complete virtual synchrony model, if the programmer desires. For

example, Horus' modules can implement total order using a token passing layer, or casual ordering using vector clocks.

Isis has limited support for partitioning. In the event that a partition forms, dividing the large group into subgroups, only the primary group is allowed to continue operating. The primary group is selected by choosing the largest partition. Horus, on the other hand implements the extended virtual synchrony model and does not have that limitation.

2.5. TRANSIS (1992)

Transis [11] was developed as a more pragmatic approach to the creation of a distributed framework. Transis is developed with the key consideration that, while multicast is the most efficient for distributing information in a view or group of processors (as point to point quickly gives rise to N^2 message complexity, where N is the number of processes.) it can be impractical, especially over a wide area network to rely on broadcast to deliver messages. Transis, then considers two components for the network: a local area component and a wide area component.

The local area component, Lansis, is responsible for the exchange of messages across a LAN. Transis uses gateways, called Xporters, to deliver messages between the local views. Transis uses a combination of acknowledgements, which are piggybacked on regular messages to identify lost messages. If a process observes a message being acknowledged that it does not receive then it sends a negative acknowledgement broadcast informing the other members of the view it did not receive that message. In Lansis, the acknowledgement signals that a process is ready to deliver a message. Lansis assumes that all messages can be casually related in a global, directed, acyclic graph and there are a number of schemes that deliver messages based on adherence to that graph.

2.6. TOTEM (1996)

Totem[12] is designed to use local area networks, connected by gateways. The local area networks use a token passing ring and multicast the messages, much like Isis and Horus. The gateways join these rings into a multi-ring topology. Messages are first delivered on the ring which they are sent, then forwarded by the gateway which connects the local ring to the wide area ring for delivery to the other rings. The message protocol gives the message total ordering using a token passing protocol. Topology changes are handled in the local ring, then forwarded through the gateway where the system determines if the local change necessitates a change in the wide area ring. Failure detection is also implemented to detect failed gateways.

2.7. SPREAD

Spread[13] is a modern distributed framework. It is designed as a series of daemons which communicate over a wide area network. Processes on the same LAN as the daemons connect to the daemon directly. This is analogous to the gateways in the other distributed toolkits, however, the daemon is a special, dedicated message passing process, rather than a participating process with a special role.

Using a dedicated daemon allows a Spread view to reconfigure less frequently when a process stops responding (which normally correlates to a view change) since the daemons can insert a message informing other processes that a process has left while still maintaining the message ordering without disruption. Major reconfigurations only need to occur when a daemon is suspected of failing.

Spread implements an extended virtual synchrony model. However, message ordering is performed at the daemon level, rather than at the group level. Total ordering is done using a token passing scheme among the daemons.

2.8. FAILURE DETECTORS

Failure detectors [14] (Sometimes referred to as unreliable failure detectors) are special class of processes in a distributed system that detect other failed processes. Distributed systems use failure detection to identify failed processes for group management routines. Because it isn't possible to directly detect a failed process in an asynchronous system, there has been a wide breadth of work related to different classifications of failure detectors, with different properties. Some of the properties include[14]:

- Strong Completeness - Every faulty process is eventually suspected by every other working process.
- Weak Completeness - Every faulty process is eventually suspected by some other working process.
- Strong Accuracy - No process is suspected before it actually fails.
- Weak Accuracy - There exists some process is never suspected of failure.
- Eventual Strong Accuracy - There is an initial period where strong accuracy is not kept. Eventually, working processes are identified as such, and are not suspected unless they actually fail.
- Eventual Weak Accuracy - There is an initial period where weak accuracy is not kept. Eventually, working processes are identified as such, and there is some process that is never suspected of failing again.

One class of failure detectors, Omega class Failure detectors, are particularly interesting because of [15]. An eventual weak failure (weak completeness and eventual weak accuracy) detector is the weakest detector which can still solve consensus. It is

denoted several ways in various works including $\diamond\mathcal{W}$ [14], \mathcal{W} [16] [17] and Ω (Omega) [15].

[15] studies an Omega class failure detector using OmNet++[18], a network simulation software package. Instead of omission failures, however, it considers crash failures. Each configuration goes through a predefined sequence of crash failures. OmNet++ is used to count the number of messages sent by each of three different leader election algorithms. Additionally, [15] only considers the system to be in a complete and active state when all participants have consensus on a single leader.

3. BACKGROUND THEORY

3.1. FREEDM DGI

In this work we model the group management module of the FREEDM DGI. The DGI is a smart grid operating system that organizes and coordinates power electronics and negotiates contracts to deliver power to devices and regions that cannot effectively facilitate their own need.

To accomplish this, the DGI software consists of a central component, the broker, which is responsible for presenting a communication interface and furnishing any common functionality needed by any algorithms used by the system. These algorithms are grouped into modules. These algorithms work in concert to move power from areas of excess supply to excess demand.

The DGI uses several modules to manage a distributed smart-grid system. Group management, the focus of this work, implements a leader election algorithm to discover which nodes are reachable in the cyber domain and to maintain group structures which are needed to migrate power.

Other modules within DGI provide additional functionality such as collecting global snapshots, and a module that negotiates the migrations and gives commands to physical components.

The DGI is a real-time system: certain actions (and reactions) involving power system components need to be completed with a pre-specified time-frame to keep the system stable. The DGI uses a round robin scheduler: each module is given a predetermined window of execution which it may use to perform its duties. When a module's time period expires, the next module in the line is allowed to execute.

3.2. BROKER ARCHITECTURE

The DGI software is designed around a broker architectural specification. Each core functionality of the system is implemented within a module that is provided access to interfaces which deliver functionality such as scheduling requests, message passing, and a framework to manipulate physical devices.

The Broker provides a common message passing interface that all modules are allowed to access. Information is passed between modules using this message passing interface. For example, the list of peers in the group is made available to other modules with a message.

Several of the distributed algorithms used in the software require the use of ordered communication channels. To achieve this, FREEDM provides a reliable ordered communication protocol (The sequenced reliable connection or SRC) to the modules, as well as a “best effort” protocol (The sequenced unreliable connection or SUC) which is also FIFO (first in, first out), but provides limited delivery guarantees.

We elected to design and implement our own simple message delivery schemes in order to avoid complexities introduced by using TCP in our system. During development, it was observed that constructing a TCP connection to a node that had failed or was unreachable took a considerable amount of time. We elected to use UDP packets which do not have those issues, since the protocol is connectionless. UDP also allows development of protocols with various properties to evaluate which properties are desirable. As a result, lightweight protocols which are best effort oriented were implemented to deliver messages as quickly as possible within the requirements.

The decision to go with a lighter weight protocol was also influenced by the FREEDM center targeting lower cost, less powerful implementation platforms, with less available computing resources than a traditional server or desktop. Furthermore, the protocols listed here continue operating despite omission failures: they follow the

assumption that not every message is critical to the operation of the DGI and that the channel does not need to halt entirely to deliver one of the messages.

3.2.1. Sequenced Reliable Connection. The sequenced reliable connection is a modified send and wait protocol with the ability to stop resending messages and move on to the next one in the queue if the message delivery time exceeds some timeout. When designing this scheme we wanted to achieve several criteria:

- Messages must be delivered in order - Some distributed algorithms rely on the assumption that the underlying message channel is FIFO.
- Messages can become irrelevant - Some messages may only have a short period in which they are worth sending. Outside of that time period, they should be considered inconsequential and should be skipped. To achieve this, we have added message expiration times. After a certain amount of time has passed, the sender will no longer attempt to write that message to the channel. Instead, he will proceed to the next unexpired message and attach a “kill” value to the message being sent, with the number of the last message the sender knows the receiver accepted.
- As much effort as possible should be applied to deliver a message while it is still relevant.

There one adjustable parameter, the resend time, which controls how often the system would attempt to deliver a message it hadn't yet received an acknowledgement for. There is a `Resend()` function is periodically called to attempt to redeliver lost messages to the receiver.

3.2.2. Sequenced Unreliable Connection. The SUC protocol is simply a best effort protocol: it employs a sliding window to try to deliver messages as quickly as possible. A window size is decided, and then at any given time, the sender

can have up to that many messages in the channel, awaiting acknowledgement. The receiver will look for increasing sequence numbers, and disregard any message that is of a lower sequence number than is expected. The purpose of this protocol is to implement a bare minimum: messages are accepted in the order they are sent.

Like the SRC protocol, the SUC protocol's resend time can be adjusted. Additionally, the window size is also configurable, but was left unchanged for the tests presented in this work.

3.2.3. Real Time. The DGI's specifications also call for real time reaction to events in the system. The DGI's real-time requirements are designed to enforce a tight upper bound on the amount of time used creating groups, discovering peers, collecting the global state, and performing migrations.

To enforce these bounds, the real-time DGI has distinct phases which modules are allowed to use for all processing. Each module is given a phase which grants it a specific amount of processor time to complete any tasks it has prepared. When the allotted time is up the scheduler changes context to the next module. This interaction is shown in Figure 3.1

Modules inform the scheduler of tasks they wish to perform by either submitting them to be performed at some point in the future, or informing the scheduler of a tasks that is ready to be executed immediately.

Tasks become ready, either by being inserted as ready, or the time period that specified when it should be executed after has passed. The prepared task is inserted into a ready queue for the module that the task has been scheduled for.

When that modules phase is active, the task is pulled from the ready queue and executed. When the phase is complete, the scheduler will stop pulling tasks from the previous modules queue and begin pulling from the next modules queue.

This allows enforcement of upper bound message delay. The modules have a specific amount of processing time allotted. Modules with messages that invoke

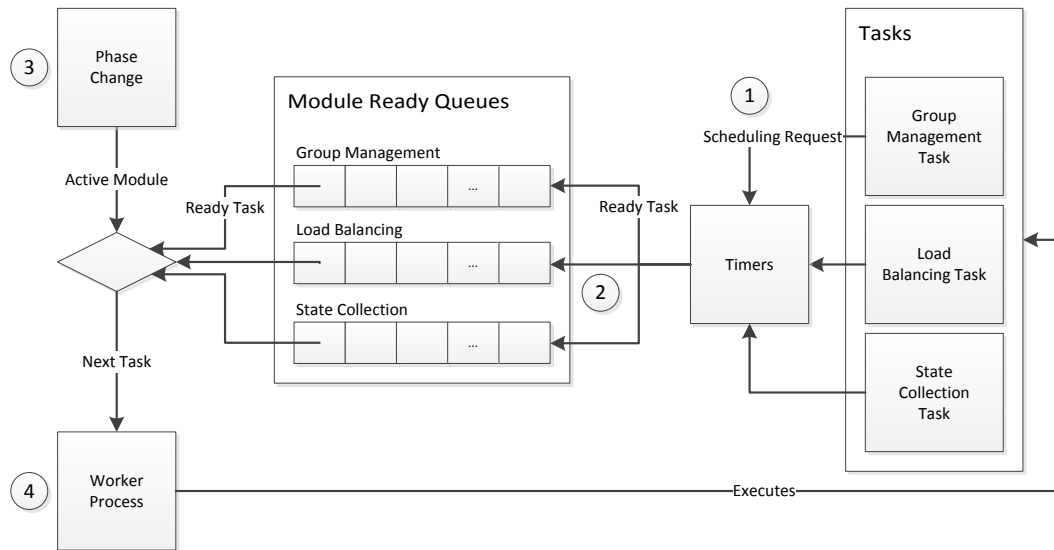


Figure 3.1: The real time scheduler uses a round robin approach to allot execution time to modules.

1. Modules request that a task be executed by specifying a time in the future to execute a task. A timer is set to count down to the specified moment. Modules may also place tasks immediately into the ready queue if the task may be executed immediately.
2. When the timer expires the task is placed into the ready queue for the module that requested the task be executed.
3. Modules are assigned periods of execution (called phases) which are a predetermined length. After the specified amount of time has passed, the module's phase ends and the next module in the schedule's tasks begin to execute.
4. The worker selects the next ready task for the active module from the ready queue and executes it. These tasks may also schedule other tasks to be run in the future.

responses (or a series of queries and responses) typically are required to be received within the same phase, using round numbers which enforce that the message was sent within the same phase.

Modules are designed and allotted time to allow for parameters such as maximum query-response time (based on the latency between communicating processes).

This implies that a module which engages in these activities has an upper-bound in latency before messages are considered lost.

3.3. GROUP MANAGEMENT ALGORITHM

The DGI uses the leader election algorithm, “Invitation Election Algorithm” written by Garcia-Molina in [19]. Originally published in 1982, his algorithm provides a robust election procedure which allows for transient partitions. Transient partitions are formed when a faulty link between two or more clusters of DGIs causes the groups to temporarily divide. These transient partitions merge when the link is more reliable. The election algorithm allows for failures that disconnect two distinct sub-networks. These sub networks are fully connected, but connectivity between the two sub-networks is limited by an unreliable link.

Since Garcia-Molina’s original publication, there has been a large body of work creating various election algorithms. Each algorithm is designed to be well suited to the circumstances it will be deployed in: there are specialized algorithms for wireless sensor networks[20][21], detecting failures in certain circumstances[22][23], and transient partitions. Work on leader elections has been incorporated into a variety of distributed frameworks: Isis[8], Horus[10], Totem[12], Transis[11], and Spread[13] all have methods for creating groups. Despite this wide body of work, the fundamentals of leader election are consistent: nodes arrive at a consensus of a single peer who coordinates the group, and nodes that fail are detected and removed from the group.

The elected leader is responsible for making work assignments and identifying and merging with other coordinators when they are found, as well as maintaining an up-to-date list of peers for the members of his group. Likewise, members of the group can detect the failure of the group leader by periodically checking if the group leader is still alive by sending a message. If the leader fails to respond, the querying

node will enter a recovery state and operate alone until they can identify another coordinator to join with. Therefore, a leader and each of the members maintains a set of processes which are currently reachable, which is a subset of all known processes in the system.

The leader and members have separate roles to play in the failure detection process. Leaders use a periodic search to locate other leaders in order to merge groups. This serves as a ping / response query for detecting failures in the system. It is also capable of detecting a change in state either by network issue or crash failure that causes the process being queried to no longer consider itself part of the leader's group. The member will only suspect the leader, and not the other processes. Of course, simple modifications could allow the member to suspect other members by use of a heart beat or query-reply system, but it is not implemented in DGI code.

3.4. NETWORK SIMULATION

Network unreliability is simulated by dropping datagrams from specific sources on the receiver side. Each receiver was given an XML file describing the prescribed reliability of messages arriving from a specific source. The network settings were loaded at run time and could be polled if necessary for changes in the link reliability.

On receipt of a message, the broker's communication layer examines the source and randomly selects, based on the reliability prescribed in the XML file, whether or not to drop a message. A dropped message is not delivered to any of the sub-modules and was not acknowledged by the receiver. Using this method we were able to emulate a lossy network link but not one with message delays.

3.5. MARKOV MODELS

Markov models are a common way of recording probabilistic processes that can be in various states which change over time. A Markov model is a directed graph composed of states, and transitions between these states. Each transition has some probability attached to them.

Since the system, taken as a whole, can be reasonably modeled as a collection of states each describing the state or configuration of the system, and that the transitions between those state (failure events or election events) are probabilistic, rather than deterministic, it is a natural extension to model the distributed system as a Markov Chain.

Models can be expressed either in discrete time or continuous time. In a discrete time model time is divided into distinct slices. After each “slice” the system transitions based on the random chance from each of possible transitions. The discrete model also allows for a transition that returns to the same state. To contrast, a continuous time model assumes that the time between transitions are exponentially distributed.

3.5.1. Continuous Time Markov Chains. The model begins in some initial state, and then transitions into other states based on the probabilities assigned on each edge of the graph. Each state is memoryless, meaning that the history of the system, or the previous states have no effect on the next transition that occurs. Letting $F_X(s)$ equal the complete history of a Markov chain X up to the state s , and $j \in S$, where S is the complete set of states in the model, then the probability of transitioning to the next state j ($X_{n+1} = j$) only depends on the current state s (X_n): [24]

$$P\{X_{n+1} = j|F_X(s)\} = P\{X_{n+1} = j|X_n = s\} \quad (3.1)$$

Additionally, the models we present are time homogeneous, meaning that the time that transition probabilities are not affected by the amount of time that has passed in the simulation. Let $X(s) = i$ indicate the model is in state i at time s if the model is time homogenous, then the probability of transitioning to state j only depends on the time spent in that state (t).

$$P\{X(s+t) = j | X(s) = i\} = P\{X(t) = j | X(0) = i\} \quad (3.2)$$

Each transition has some expected value or holding time which describes the amount of time before a transition occurs. Continuous time models do not have transitions that return to the same state since the expected value of the transition time describes when how long the system remains in the same state. The probability density function (PDF) of the exponential distribution can be written as: [25]

$$f(x; \lambda) = \begin{cases} \lambda e^{-\lambda x} & x \geq 0 \\ 0 & x < 0 \end{cases} \quad (3.3)$$

As a result, the expected or mean value of an exponential distribution, is a function of the parameter λ : [25]

$$E[X] = \frac{1}{\lambda}. \quad (3.4)$$

When there are multiple possible transitions from a state, each with their own expected transition time, the expected amount of time in the state is: [24]

$$\sum \lambda(x, y) = \sum \lambda p_{x,y} = \lambda(x) \quad (3.5)$$

where $\lambda(x, y)$ is the expected amount of time before state x transitions to state y . Interestingly, the expected time in a state ($\lambda(x)$) is related to the expected time for an individual transition ($\lambda(x, y)$) by a probability $p_{x,y}$.

Each transition lends to an expected amount of time in the state. Then, to do a random walk of a continuous time Markov chain, an intensity matrix must also be generated in order to describe which transition is taken after the exponentially distributed amount of time in the state has passed. Consider then two streams of random variables, one of which is exponentially distributed and used to determine the amount of time in a state. The second stream is normally distributed and used to determine which state to transition to through the intensity matrix.

3.5.2. Constructing The Markov Chain. Consider a set of processes, that are linked by some packet based network protocol. In our experiments we provide two protocols, each with different delivery characteristics. Under ideal conditions a packet sent by one process will always be delivered to its destination. Without a delivery protocol, as soon as packets are lost by the communication network, the message that it contained is lost forever. Therefore to compensate for the network losing packets, a large variety of delivery protocols have been adapted. Each protocol has a different set of goals and objectives, depending on the application.

Keeping in mind that a single lost packet does not necessitate the message it contained is forever lost, different protocols allow for different levels of reliability despite packet loss.

The leader election algorithm is centered around two critical events: checking, and elections. The check system is used to detect both failures and the availability of nodes for election. Processes in the system occasionally exchange messages to determine if the other processes have crashed, and to discover new leaders.

The DGI can perform work assuming that it is in a group, and not in an election state (since the group management module instructs other modules to stop

during an election). The collected data in the previous sections is based on that assumption, and the Markov chains that models those scenarios needed to as well.

Processes in the DGI are either members or leaders. Leaders are processes that have won elections among its members.

As stated previously, it was assumed that the arrival times of events in the distributed system were distributed exponentially. Events are modeled in the chain using $\lambda(x)$ which is the parameter of the exponential distribution.

Failure Detection. When a leader sends its check messages, the nodes that receive it either respond in the positive, indicating that they are also leaders, or in the negative indicating that they have already joined a group. This message is sent to all known nodes in the system. If a process replies that it is also a leader, the original sender will enter election mode and attempt to combine groups with the first process. Nodes that fail to respond are removed from the leader's group, if they were members.

The member on the other hand will only direct its check message to the leader of its current group. As with the leader's check message, the response can either be positive or negative. A "yes" response indicates that the leader is still available and considers the member a part of its group. A "no" response indicates that either the leader has failed and recovered, or it has suspected the member process of being unreachable (either due to crash or network issue) and has removed them from the group. In this event the member will enter a recovery state and reset itself to an initial configuration where it is in a group by itself.

On any membership change, either due to recovery, or a suspected failure, the list of members for a group is pushed to every member of that group by the leader. Members cannot suspect other processes of being crashed, only the leader can identify failed group members.

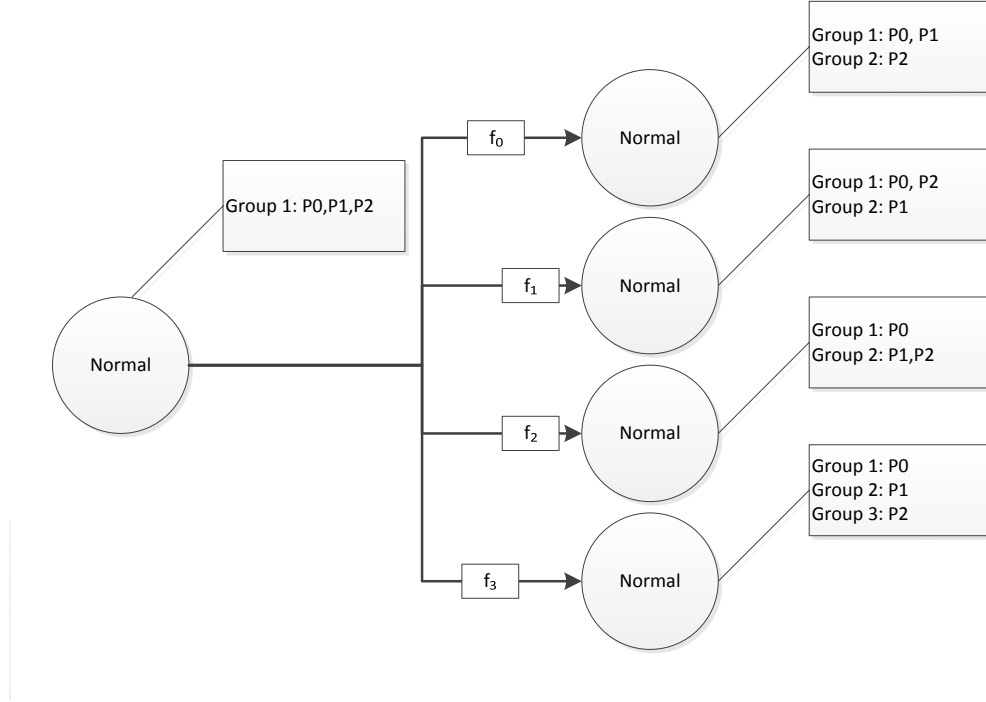


Figure 3.2: A diagram showing a partial Markov chain for failure detection

A model of a failure detection stage of the leader election algorithm is presented in Figure 3.2. A set of nodes begin in a normal state as part of a group. The leader sends a query to every member, and every member sends a query to the leader. If a response is not received in either direction, the process is considered to be unreachable and is either ejected from the group by the leader (if the query originated from the leader) or the member leaves the group and becomes a coordinator themselves.

The system will stay in the original state as long as all nodes complete their queries and responses. Let T_R be the amount of time allowed for a response, T_C be the time between discovery attempts, and p_F is the probability that at least one peer fails to complete the exchange. Based on this, the expected amount of time in the grouped state (T_G) is:

$$\begin{cases} T_G = (T_R + T_C)/p_F & p_F > 0 \\ \infty & p_F = 0 \end{cases} \quad (3.6)$$

Let λ_G equal exponential parameter of the exponential distribution for a grouped state. Then we can relate the probabilities of each possible transition to the parameter for that state. Let p_i be the probability of transitioning to configuration i after leaving the base state and let f_i be the exponential parameter for the transition to that configuration:

$$\lambda_G = \sum f_i = \sum \lambda_G p_i = \frac{1}{T_G} \quad (3.7)$$

Leader Election. During elections, a highest priority leader (decided by its process id) will send invites to the other leaders it has identified. If those leaders accept the highest priority leader's invites, they will reply with an accept message and forward the invite to their members, if there are any. If the highest priority process fails to become the leader the next highest will send invites after a specified interval has passed.

Therefore, the membership of the system can be affected in two ways: election events which change the size of groups and failure suspicion (via checks) which decreases the size of groups. Note that elections can decrease the size of groups as well as increase them: If a round of forwarding invites fails by the new leader to his original group, the group size could decrease.

When a process is initialized it begins in the "solo" state: it is in a group with itself as the only member. As nodes are discovered by checks, the processes combine into groups. Groups are not limited by increasing one a time; they can increase by combined size of the groups of the leader processes.

We define a metric to assess the performance of the system under duress: we first consider that the distributed can only perform meaningful work when the processes can work together to perform physical migration. This means that there are two networks that affect the system’s ability to do work, the physical flow network and the cyber communication network.

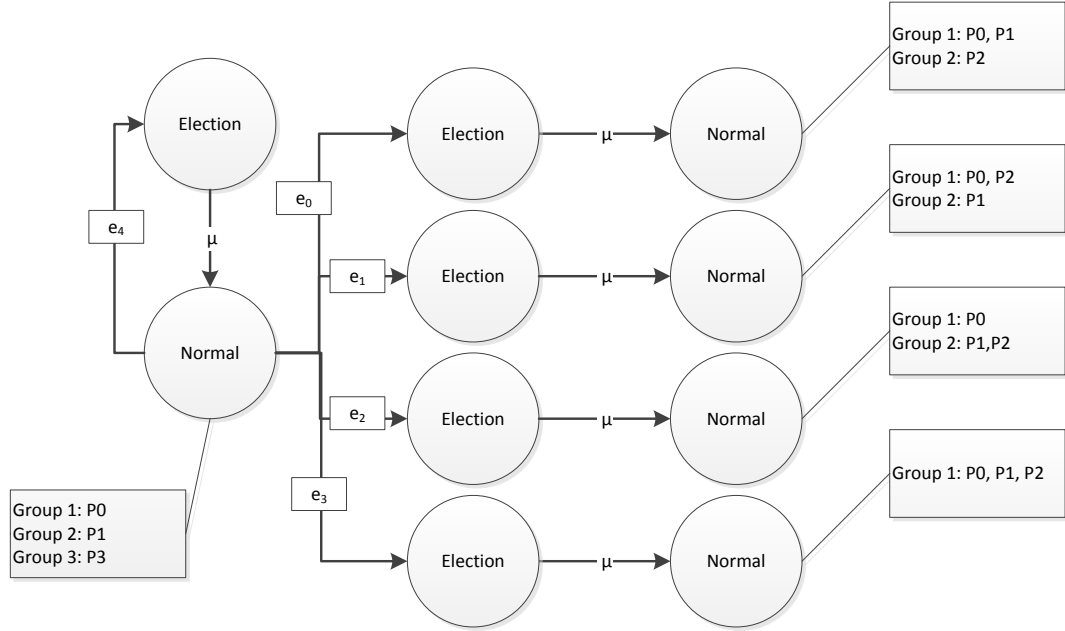


Figure 3.3: A diagram showing a partial Markov chain for an election

A continuous time Markov model of a single election is presented in Figure 3.3. A set of leaders begin in a normal state. After some time T_D an “are you coordinator” message discovers some other peer. T_D is a function of the number of discovery checks which discover no leaders (which in turn is a function of the link reliability). Let T_R be the amount of time allowed for a response, T_C be the time between discovery attempts, and p_D is the probability that the exchange discovers a leader.

$$\begin{cases} T_D = (T_R + T_C)/p_D & p_D > 0 \\ \infty & p_D = 0 \end{cases} \quad (3.8)$$

Then, the parameters e_x in Figure 3.3 are a function of T_D and p_x , the probability an election results in configuration x .

$$e_x = \frac{p_x}{T_D} \quad (3.9)$$

Once a leader has been discovered, the system transitions into an election state, based on the potential outcome, where the peers hold an election to determine a new configuration. As shown in Figure 3.3, an election can either succeed or fail, resulting in a new system configuration, or each involved process returning to the single member group state.

The amount of time that an election takes is fixed before the algorithm is executed. Let T_E be the mean time it takes to complete any election. Therefore:

$$\mu = \frac{1}{T_E} \quad (3.10)$$

Combined Model. A combined model combines election and failure detection Markov chain components. Except for the states where all reachable nodes are in the same group and the states where there are no reachable leaders each state has a combination of election transitions and failure transitions. The combined model is predictive of the overall characteristics of the system. The time spent in a particular configuration is a function of the λ 's of all the events that can cause the system to transition away from a configuration.

To construct the Markov chain, simulations of individual events are performed. The circumstances for the events are assumed to be homogeneous: processes only differ by their process id. Using this assumption, the simulation of events can be broken

down into a series of scenarios that are representative of the events in the system. Since each scenario is independent of other scenarios, each scenario can be run independently. Additionally, since the circumstances are assumed to be homogeneous, scenarios that are similar, such as ones where two processes swap roles can be simulated only once, and the results can be transformed from one scenario to another with a simple mapping. This mapping scheme and parallizability helps keep the state space explosion of the potential states under control.

3.5.3. Assumptions. In order to model the system, we assumed that the time between events was exponentially distributed. Furthermore, we assumed that the system would be fairly well synchronized, with most elections occurring at the same time. This assumption was valid for 2-Node cases of our non-real-time code, but was a major issue as the number of nodes began to increase. However, with the use of the round-robin scheduler with synchronization to enforce our real-time requirements, assuming the synchronization of processes is not a major leap.

All participating peers are assumed to be on the same schedule: all peers begin execution of a model simultaneously. This is accomplished using [26]. This work assumes that the clocks are synchronized: even if the network has partitioned, process clocks have not drifted noticeably from their last synchronization. Additionally, a production system would likely use GPS time synchronization in order to take certain power system readings [27].

4. RESULTS

4.1. INITIAL RESULTS

Initial data was collected from a non-real time version of the DGI code. The collected results from the tests are divided into several target scenarios as well as the protocol used.

The first minute of each test in the experimental test is discarded to remove any transients in the test. The result is that while the tests were run for ten minutes, the maximum result is 9 minutes of in-group time. These graphs first appeared in [28].

4.1.1. Sequenced Reliable Connection.

Two Node Case. The 100ms resend SRC test with two nodes can be considered a sort of a control. These tests, pictured in Figure 4.1, highlight the performance of the SRC protocol, achieving the maximum in-group time of 9 minutes with only 15% of datagrams arriving at the receiver.

Figures 4.2 demonstrates that as the rate at which lost datagrams are re-sent is decreased to once every 200ms (from 100ms) the time in group falls off. This behavior is expected, since each exchange has a time limit for each message to arrive and the number of attempts is reduced by increasing the re-send time.

Transient Partition Case. The transient partition case shows a simple example where a network partition separates two groups of DGI processes. In the simplest case where the opposite side of the partition is unreachable, nodes will form a group with the other nodes on the same side of the partition. In our tests, there are two nodes on each side of the partition. In the experiment, the probability of a

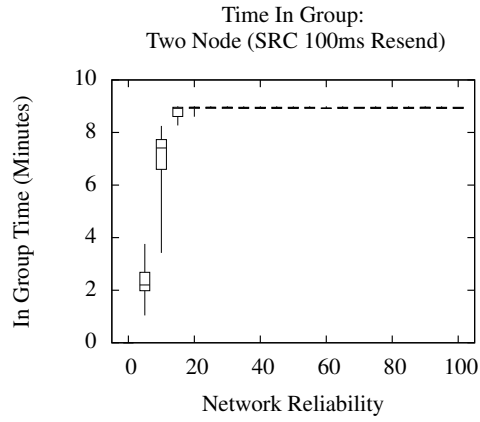


Figure 4.1: Time in group over a 10 minute run for two node system with 100ms resend time

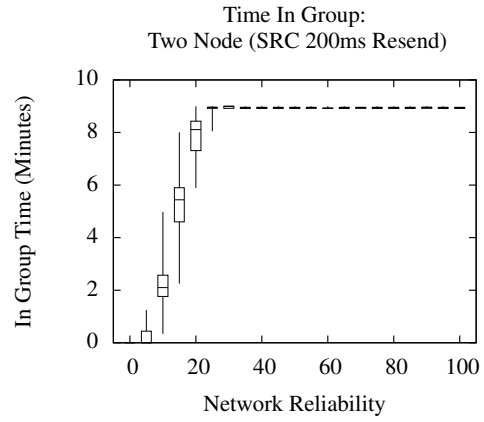


Figure 4.2: Time in group over a 10 minute run for two node system with 200ms resend time

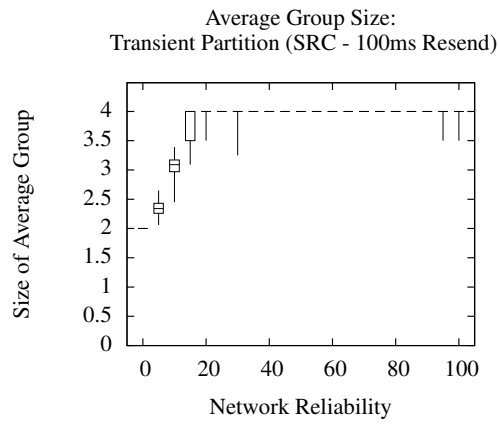


Figure 4.3: Average size of formed groups for the transient partition case with 100ms resend time

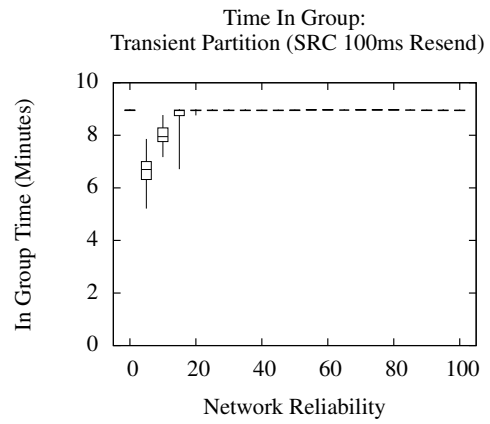


Figure 4.4: Time in group over a 10 minute run for the transient partition case with 100ms resend time

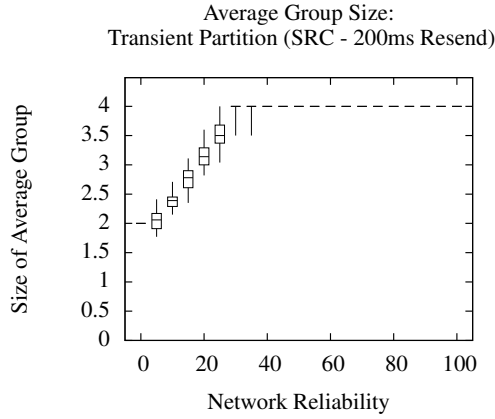


Figure 4.5: Average size of formed groups for the transient partition case with 200ms resend time

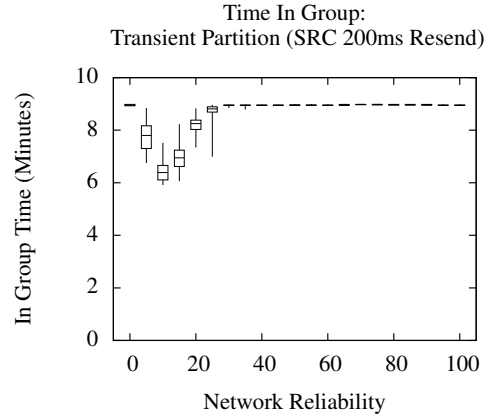


Figure 4.6: Time in group over a 10 minute run for the transient partition case with 200ms resend time

datagram crossing the partition is increased as the experiment continues. The 100ms case is shown in Figures 4.3 and 4.4.

While messages cannot cross the partition, the DGIs stay in a group with the nodes on the same side of the partition leading to an in group time of 9 minutes, the maximum value. As packets begin to cross the partition (as the reliability increases), DGI instances on either side begin to attempt to complete elections with the nodes on the opposite partition and the time in group begins to fall. However during this time, the mean group size continues to increase, meaning while the elections are decreasing the amount of time that the module spends in a state where it can actively do work, it typically does not fall into a state where it is in a group by itself, which means that most of the lost in group time comes from elections.

The 200ms case, shown in Figures 4.5 and 4.6 displays similar behavior, with a wider valley due to the limited number of datagrams. It is also worth noting that the mean group size is less than 2 for some reliability levels in the figure, possibly because the longer resend times allow for more race conditions between potential

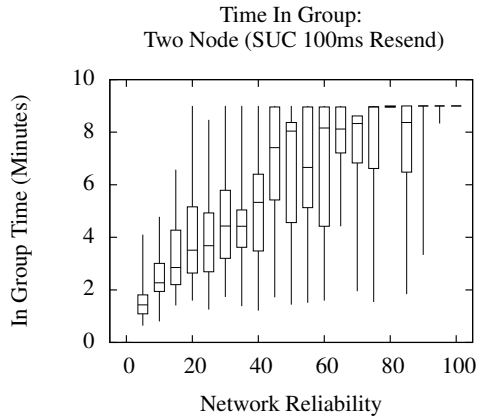


Figure 4.7: Time in group over a 10 minute run for two node system with 100ms resend time

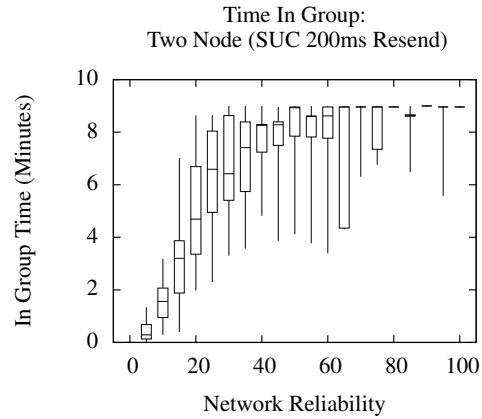


Figure 4.8: Time in group over a 10 minute run for two node system with 200ms resend time

leaders. These issues are more pronounced in the SUC scenarios and are discussed in that section.

4.1.2. Sequenced Unreliable Connection.

Two Node Case. The SUC protocol's experimental tests show an immediate problem: although there is a general trend of growth in the amount of time in group and group size charts, shown in Figure 4.7 there is a high amount of variance for any particular trial.

In the 200ms resend case, shown in Figure 4.8, it can be observed that there is a more growth rate in the in-group time as the reliability increases. In fact, averaging across all the collected data points from the experiment, the average in-group time is higher for the 200ms case than it is for the 100ms case (6.86 vs 6.09). However, due to the large amount of variance in the collected in-group time, it is not possible to state with confidence that there is a significant difference between the two cases.

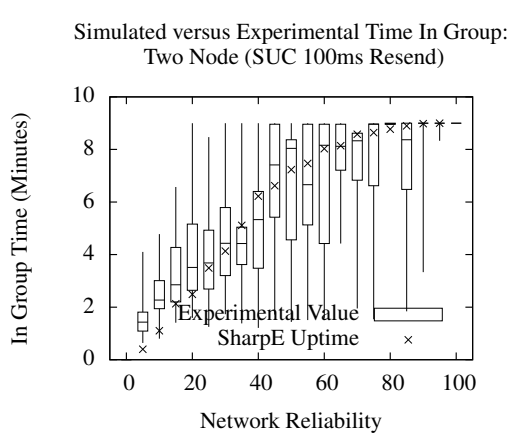


Figure 4.9: Comparison of in-group time as collected from the experimental platform and the simulator (1 tick offset between processes).

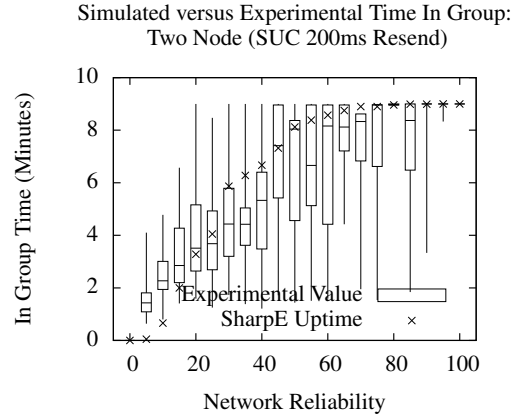


Figure 4.10: Comparison of in-group time as collected from the experimental platform and the simulator (2 tick offset between processes).

4.2. MARKOV MODELS

Due to the high amount of variance in the collected data, and the resulting difficulty making any sort of prediction about other systems from the data, a more formal approach was tried.

4.2.1. Initial Model Calibration. The presented methodology of constructing the model was initially calibrated against the original two-node case, using a non-real-time version of the DGI codebase. The resulting Markov chain was processed using SharpE [29][30] made by Dr. Kishor Trivedi's group at Duke University, a popular tool for reliability analysis. SharpE measured the reward collected in 600 seconds, minus the reward that was collected in the first 60 seconds (to emulate that the first 60 seconds were discarded in the experimental runs.) The SharpE results are plotted along with the experimental results in Figures 4.9 and 4.10.

The race condition between processes during an election is a consideration in the original leader election algorithm, and is an additional factor here. The simulator

provided a parameter to allow the operator to select how closely synchronized the peers were (the time difference between when each of them would search for leaders.) The exchange of messages, particularly during an election had a tendency to synchronize nodes during elections, and so the nodes could synchronize even if they did not initially begin in a synchronized state. As a result, the simulation results aligned best for the 100ms resend case with 1 ticks (Approximately 100ms difference in synchronization between processes) and 2 ticks (Approximately 400ms) in the 200ms resend case.

Models fit to the non-real-time code in groups larger than 2 processes did not fit well. This is presumed to be a combination of several factors. The suspected major source of error included the structure of the chain, which naturally assumes that all processes enter the election state a roughly the same time, which is not typically true for any number of processes greater than 2. Additionally, the simulator could only assume that the synchronization between processes was mostly fixed, which was not the case in the larger configurations, since the coincidental synchronization that occurred in the two node case was suppressed by the increased number of peers. Furthermore, an issue with SharpE was discovered that prevented the particular structure of the chains produced from being handled correctly. In particular, the structure of elections, where a chain goes through election state to a normal state causes the semi-symbolic solver SharpE uses to produce an unreasonable expected reward. To circumvent that issue, SharpE was replaced by a random-walker which generates exponentially distributed numbers and follows the paths of the chain, across several hundred trials, in order to collect time in group data for models which SharpE cannot process.

The structure of the Markov Chain, which assumed that processes enter the election state mostly simultaneously was an appropriate assumption for the real-time system, since the round-robin scheduler synchronizes when processes run their group

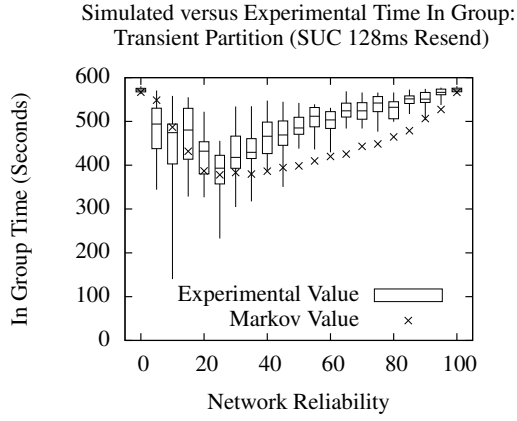


Figure 4.11: Comparison of in-group time as collected from the experimental platform and the time in group from the equivalent Markov chain (128ms between resends).

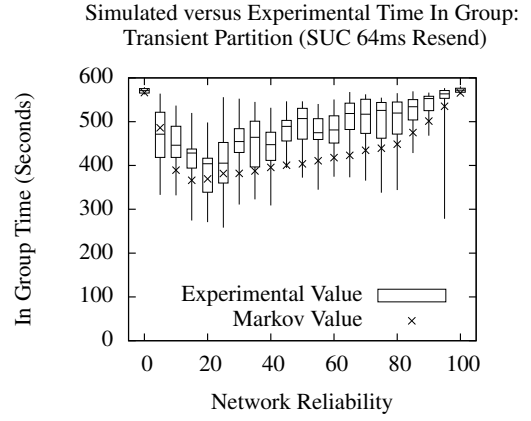


Figure 4.12: Comparison of in-group time as collected from the experimental platform and the time in group from the equivalent Markov chain (64ms between resends).

Table 4.1: Error and correlation of experimental data and Markov chain predictions

Re-send	Correlation	Error
128	0.7656	11.61%
64	0.8604	11.70%

management modules. The simulator was set to assume that the synchronization between processes was less than one resend interval, and new experimental data was collected for the 4 node, transient partition case. The collected data is overlaid with the results from the random walker in Figures 4.11 and 4.12.

As a measure of the strength of the model, the correlation with the predicted value was compared. The average error was also computed for each of the samples taken. This information is presented in Table 4.1.

5. CONCLUSION

This work presented a new approach for predicting the behavior of a real-time distributed system under omission failure conditions. By using a continuous time Markov chain, a variety of insights can be gathered about the system, including observations such as how long a particular configuration will be stable, and the behavior of the system in the long run. The Markov results will be used to make better real time schedules to better react to the network faults we plan on introducing to our test-beds. The primary concern are scenarios in which the cyber controller attempts to make physical components which are not connected in the physical network interact, and scenarios where a fault in the cyber network causes the paired events (where two physical controllers change to accomplish some transaction or exchange) to only be partially executed. For example, in the DGI load balancing scheme, a node in a supply state injects a quantum of power into the physical network, but the node in the demand state does not change to accept it. These errors, which are the primary focus of this work could cause instability if a sufficient number of these failed exchanges occur. In [6], Choudhari et. al. show that failed transactions can create a scenario where the frequency of a power system could become unstable.

Moving forward, we have identified these areas as targets for improving the research done and creating new contributions.

5.1. SELECTING A SCHEDULE

As shown previously, there is a relationship between congestion and the amount of time a process spends in a group. We have shown that an exponential distribution as part of a continuous-time Markov chain can be used to describe the expected time in a configuration given an omission fault rate. Based on this, we expect to be able

to develop a function which relates the congestion in the network to predict how long a pair of processes will be able to interact. Therefore, when a process observes congestion, either from the network hardware or lost messages they will be able to produce an estimate of their ability to stay in a group with a process. If we consider a set of processes that are in a group together then there is some function which describes the expected amount of time until the group would need to reconfigure:

$$E(c)[X] = \begin{cases} 0 & \text{if } c = 1.0 \\ \infty & \text{if } c = 0.0 \\ 1/\lambda(c) & \text{otherwise} \end{cases} \quad (5.1)$$

Where c is a measure of the congestion in the system: the expected time in group is zero when no messages can be delivered, and infinite when no messages are lost. As congestion is observed by processes, they produce new estimates of λ as a function of the congestion c . The value of $\lambda(c)$ may be based on a series of collected data sampled discretely (using simulation) or may be mappable with a probability distribution.

Additionally, congestion can be used to predict how many migrations will fail. Given a real time schedule and a congestion value c , as with the groups we can define a function $F(c)$ describes the likelihood of a migration failing. These failed migrations contribute to a k -value which can be used in a simplified version of the physical invariant from [6][31][32].

$$\{k < \text{max_outstanding_migrations}\} \quad (5.2)$$

If the number of failed migrations k exceeds the maximum number of outstanding migrations the system will be unstable. A CPS implementing this invariant will

Table 5.1: Comparison of two proposed schedules, A and B

Schedule A		Schedule B
$E_A(c)[X]$	$<$	$E_B(c)[X]$
$F_A(c)$	$>$	$F_B(c)$
R_{Am}	$>$	R_{Bm}

stop performing migrations when this invariant would be violated. We can define a related value *remaining_k*:

$$remaining_k = max_outstanding_migrations - k \quad (5.3)$$

Then we can produce a relationship of when a group is likely to violate the physical invariant in a round, given an upper bound on the number of migrations that will be attempted that round:

$$\{F(c) * max_migrations \geq remaining_k\} \quad (5.4)$$

Given this, we can then consider the time that a process may live for, the time it would take to reconfigure, and the likelihood of violating the physical invariant into account as part of selecting a real-time schedule. For example, given that it maybe easier to maintain an existing group rather than elect a new one during network congestion, we can select a new schedule that optimizes the amount of work done and the health of the physical system. Consider a pair of real-time schedules A and B. Schedule A favors high-performance: it expects very little congestion on the network, in exchange however, it can complete many more migrations per round (Noted as R_m). Schedule B is slower and safer. It has longer timeout periods that allow for more omission failures. However, as a result, it can't do a much work: the rate that migrations are performed is lower. The relationship between schedule A and B in terms of the functions $E(c)[X]$ and $F(c)$ in summarized in Table 5.1.

If we define a function $migrations(R_m, t)$ that relates a time period t and the rate migrations are attempted to a number of attempted number of migrations over a time period t . With this function a relationship between the number of failed migrations on schedule A to the lower amount of work produced on schedule B, assuming the migration size is constant:

$$\Delta(A, B) = (F_A(c) * migrations(R_{Am}, t)) - (F_B(c) * migrations(R_{Bm}, t)) \quad (5.5)$$

Based on the above equation, schedule B should be selected when Δ is negative, yielding the invariant for the current schedule x and the set of potential schedules S , which validates when is best to apply a given schedule in a group:

$$\{\Delta(x, y) \geq 0, \forall y \in S\} \quad (5.6)$$

The rate that the system should reconfigure is a function of the maximum number of failed migrations that the system can handle before becoming unstable, the time it takes to write to the channel and the time it takes process messages. The amount of time in group can also be a consideration for which algorithm to select based on the needed amount of time to perform its work. Group Management can be used as a critical component in a real-time distributed system to manage the number of lost messages and as a consequence, the number of failed migrations in a CPS. It is critical to understand how frequently nodes enter and exit the group based on lost messages and how many migrations fail as a consequence of those messages. This area is deficient because it is strongly coupled to the interactions with the physical component: we must understand how the cyber configuration and physical changes made by that configuration can affect the system, and establish when reconfigurations should occur to keep the system stable.

5.2. CORRECTNESS OF AN INSTALLED CONFIGURATION

The work presented in this document is probabilistic: the results of a leader election are random and based only on responses arriving within a specified period of time. Other factors can affect what configurations can be installed such as trust in the parties in the group, the underlying physical topology, and the reliability of the peers in that group. We will develop guards on the properties of a configuration that protect the physical topology and the members of the group. These guarantees would also allow processes to better police the configurations they are installed in, in order to protect the system from malicious nodes.

These guards can ensure quantities like trust in the involved properties, the physical organization for verification methods like attestation. Guards could also ensure the capability of the group to do work: a formed group in which all processes are in supply or demand can do no work. Likewise, if the congestion is too high between a pair of processes, it will affect the ability of the group to do work. Perhaps most importantly, we wish to ensure that a partition in the cyber domain will not cause a connected physical topology to become unstable. Interference between groups in the physical domain should not allow a global physical invariant (P_{IG}) to be violated. We wish to develop guards that ensure when all local invariants are true the global physical invariant is true, where P_{Ig} is the physical invariant for a group g :

$$\bigwedge_{g \in Groups} P_{Ig} \rightarrow P_{IG} \quad (5.7)$$

5.3. ACCURACY AND SCOPE OF THE MODEL

We expect to be able to further refine the model and the algorithms and formulas for generating the model. There are some features of the behavior of the

DGI which are not completely encapsulated in the model. Currently, the arrival times are a continuous time estimation of events that occurs discretely in the real-time system. The failure detection checks occur on a specific interval, but the continuous time Markov Chain allows these events to occur at any moment. This limitation reduces the accuracy of the model.

With respect to the way the models are generated with the simulator, the way models are specified can be generalized to support more systems of similar design. Additionally, the simulator uses the resend interval as the smallest timestep, which allows models to be evaluated quickly, but limits accuracy. Lastly, it is worthwhile to adapt the simulation to use a more common network simulation software like OmNet++ to make the results more portable.

The models presented in this work focus only on the leader election component of a dynamically configured CPS. Additional work would incorporate additional components of the DGI system into the models for a more complete picture of the behavior of the system during failures. We will consider the correctness of the incorporated algorithms, how omission failures can violate that correctness, and what restrictions we can place on the configuration and operation of DGIs in order to protect the entire system during failures. To do this, we will expand the analysis performed here to incorporate algorithms such as state collection and load balancing and define metrics to quantify their behavior during omission failures. This thrust will pair with the correctness and time between reconfigurations: different algorithms will have different amounts of failure that can be allowed before reconfiguration is necessary.

5.4. DELIVERABLES

Therefore, moving forward, we will expand the models we have presented here to include more of the properties of the complete CPS. This model will allow us to better understand what effects the group behavior has on the CPS. Using this, we can establish invariants which allow us to ensure the correctness of a CPS by providing assertions which will not be broken during execution. Creating these invariants will allow us to improve the development of CPSs, especially in their dynamic configuration, which is an area with limited development. These invariants also allow us to create an assertion of correctness which can be validated, during runtime, to ensure the system maintains its stability. We will continue to create and validate models of the CPS against simulations and actual hardware. As we do so we will construct invariants that describe the correct behavior of the groups to ensure safe operation. In addition to the existing conference paper [28], we have prepared a journal paper with the new work in this document.

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