Copyright is owned by the Author of the thesis. Permission is given for a copy to be downloaded by an individual for the purpose of research and private study only. The thesis may not be reproduced elsewhere without the permission of the Author.

A Study of a Java based Framework for Telecommunications Services

A dissertation submitted in fulfilment of the requirements for the degree of Master of Science in Computer Science Massey University, New Zealand

David Ian Ferry, 2000

Abstract

In this report, we study some of the general issues surrounding the area of telecommunications service development including the history of telecommunications services, current service creation techniques and the network used by services. We also discuss the lack of service portability and reasons for it. The JAIN framework – a set of Java APIs for integrated networks – is introduced as an approach that elegantly addresses this. We present a survey of recent work in telecommunications services that relate to JAIN. This includes a discussion of the feature interaction problem, an overview of the Telecommunications Information Networking Architecture, in particular, its relationship with JAIN, and the rapidly advancing field of Internet Telephony. In order to demonstrate the effectiveness of the JAIN framework. These services are Internet Call Waiting and Click-to-Dial. Finally, areas for future research are introduced.

Contents

1	Intr	oducti	on 1						
	1.1	Teleco	mmunication services						
		1.1.1	JAIN						
2	Background								
	2.1	Histor	y of the wired-line telephone network						
		2.1.1	The Plain Old Telephone System						
		2.1.2	Stored Program Control						
		2.1.3	Traditional Call Management						
		2.1.4	Out-Of-Band Signaling 7						
		2.1.5	The IN/1 Architecture						
		2.1.6	Advanced Intelligent Network Architecture 11						
	2.2	Teleco	mmunications Service Development						
	2.3		S7 protocol stack						
		2.3.1	The ISDN User Part						
		2.3.2	Transactional Capabilities Application Part						
		2.3.3	The Signaling Connection Control Part 23						
		2.3.4	The Message Transfer Part						
		2.3.5	The Telephone User Part						
	2.4	Java A	PIs for Integrated Networks						
		2.4.1	Introduction to JAIN						
		2.4.2	The layered model of JAIN 28						
		2.4.3	The JAIN protocol interfaces						
		2.4.4	The Call Control layer						
		2.4.5	The Security layer						
		2.4.6	Parlay						
		2.4.7	The Service Logic Execution Environment						
		2.4.8	Operations Administration and Maintenance 33						
3	Related and Previous work 35								
	3.1	Featur	e or Service Interaction						
	3.2	TINA							
		3.2.1	Conformance testing						
			이번 그만에 다 나는 것은 것이 같은 것이 같은 것이 같은 것이 있는 것이 있는 것이 있다. 이 것이 같은 것이 같은 것이 같은 것이 같은 것이 같은 것이 같은 것이 없다.						

		3.2.2	Quality of service	C
		3.2.3	Mobile telephony use	2
		3.2.4	Internet Telephony and TINA	3
	3.3	Intern	et Telephony	6
		3.3.1	Introduction	6
		3.3.2	Interworking IP and conventional telephony	7
		3.3.3	Internet Telephony Protocols	7
		3.3.4	Call Control protocols	9
		3.3.5	Other IP Telephony protocols	8
		3.3.6	Implementation of IP telephony services	1
	3.4	Other	Java-based Telephony initiatives	2
		3.4.1	Java Telephony API	2
		3.4.2	The "Softswitch"	1
4	Ser	vice ex	camples using JAIN 65	5
	4.1	Introd	luction	5
	4.2	Intern	et Call Waiting	5
		4.2.1	ICW implementation using JAIN	8
		4.2.2	The Carrier Part	2
		4.2.3	The Internet Service Provider Part	7
		4.2.4	The Call Recipient Part	0
		4.2.5	ICW security considerations	3
	4.3	Click-	to-Dial	4
		4.3.1	Introduction	4
		4.3.2	Extending the Web Browser	5
		4.3.3	A Click-to-dial architecture	6
		4.3.4	CTD operation	7
		4.3.5	CTD and JAIN)
		4.3.6	CTD security considerations	1
5	Sun		and Future Work 93	
	5.1		ary	
	5.2	Future	e Work	1
Α		-	ementation of Internet Call Waiting 97	
			luction	
		Overv		
	A.3	Displa	ying the prototype)
в	Glo	ssary	104	1
Bi	bliog	raphy	106	6

List of Tables

2.1	The SIBS in the Intelligent Network Capability Set 1	15
2.2	TCAP dialogues.	24
2.3	TCAP components.	24
3.1	The methods of SIP	52
4.1	Possible method calls for a carrier part implementation. See figure	
	4.3 for the corresponding sequence diagram	75
4.2	Possible method calls for an ISP part implementation. Refer to Fig-	
	ure 4.4 for the corresponding sequence diagram	79
4.3	Possible method calls for a call recipient part implementation. Refer	
	to Figure 4.5 for the corresponding sequence diagram	81

List of Figures

2.1	The SS7 enabled switching network. Each switch is represented by a circle. Switches were connected via two networks. The voice trunk network and the signaling network. Each node on the signaling net- work is assigned a unique SS7 address or <i>Signaling Point Code</i> (SPC)	8
2.2	The Intelligent Network 1 \ldots	10
2.3	The AIN or IN architecture. This closely resembles the IN/1 archi- tecture described earlier in Figure 2.2. However, the difference lies in the generic, standard communication between the switching system and the service control points.	11
2.4	An 800-number translation service illustrating primary components of the AIN. The messages are explained in the text.	13
2.5	An example of the development of a Call-Screening service in AIN. Each block in this diagram represents a SIB. It is to be noted that this diagram is intended to illustrate service development, rather than show the actual SIBs used in an implementation of a service.	16
2.6	The SS7 protocol stack and the ISO's OSI reference model for data communications.	20
2.7	An example of call setup and termination using the ISUP protocol. An explanation of the messages is given in the text.	21
2.8	The layered model of the JAIN approach. At the lowest level are the protocols. Above them is the call control and transactional layer. Both of these are within the secure carrier's network which can be accessed remotely via the security layer.	29
3.1	An example of call setup in SIP. See text for the message key	53
3.2	The use of MGCP integrating the PSTN and Internet Telephony	56
4.1	Internet Call Waiting GUI	67
4.2	ICW architecture using JAIN	72

4.3	A UML sequence diagram of the Carrier Part service logic that is intended to be illustrative of a possible implementation. Each num-	
	ber represents a message being passed between objects. Refer to the	
	text for an explanation of these. It is important to note that the	
	actual event ordering generated by the JainTcapProvider can differ	70
4.4	upon receiving different TCAP messages	10
4.4	Provider part. Each of the numbers in this diagram represents a message being passed between objects. These messages are explained	
	in the text.	80
4.5	A sequence diagram of a minimal end user part application that	
	is intended to be illustrative of a possible implementation. Each number represents a message being passed between objects. These	8
	messages are described in the text.	82
4.6	The association of various elements used in the CTD service	88
A.1	Communication between the simulators. Both the SSP and SCP	
	reside on the same host. The Status Monitor is not included for	
	clarity	99
A.2	The Status Monitor.	
	An incoming call notification.	

Preface

Motivation

I came across the early stages of the JAIN framework during the summer break of '98. I was attempting to find a research project that was sufficiently pragmatic as to allow me to receive funding under the GRIF program offered by the Foundation for Research, Science and Technology. I was interested in JAIN as I had previously enjoyed networking papers and systems development, and the JAIN website talked about the convergence between the Internet and traditional telephony in a standard manner. It struck me that I had no accurate idea of how the telephone network functioned. Furthermore there was very little easily accessible documentation on its workings. I was also interested in *standard* based environments such as various Unix systems, and decided that JAIN was attempting to achieve "a good thing".

The work presented in this dissertation is useful in several aspects. It presents a number of telecommunications related concepts in a single logical unit that are dispersed in many publications elsewhere. It explains the existing wired-line telephone network in depth and discusses the exciting area of Internet Telephony which we have all heard so much about, but seen so little. It *demonstrates* the value of a standard based approach by designing services and, as such, the reader may understand the framework in far greater depth than by merely reading specification documents.

Acknowledgements

First of all I would like to thank my supervisor, Dr. Anand V. Raman, for his invaluable advice during this project. Because of Anand's comments I believe that I have a far greater understanding of the scientific process, and a greater respect for academia than I had previously. Anand also helped me find a more fitting balance between an algorithm and an implementation. I am also grateful for his input as this project is not in his field of research.

I'd like to thank SolNet, especially Murray McNae, for allowing me do this project. Thanks go to the Foundation for Research Science and Technology for funding. Project funding was particularly helpful, and without it I would probably not have undertaken a Masterate degree.

Next I would like to thank the other D's that made up the three D's at SolNet – David Long and Dr. David Page. Both of these guys taught me a lot during the research and input valuable ideas and provided insightful criticism. Hopefully we can be on the receiving end of more E-gratuitous errors from our software !

I'd also like to thank Paul Lyons of Massey University for several conversations that helped me to gain a fuller understanding of research, and one on the ideal telephone. The Institute support staff were also very helpful in keeping the machines running so that the post-grad students could concentrate on our work.

The Sun JAIN team (John Dekeijzer, Doug Tait and Rob Goedman) were extremely helpful with equipment and discussion during Supercomm '99 and the July '99 JAIN conference at Telcordia. Particular thanks go to Doug Tait for getting me up on a surf board for the first time. Praise the Lord ! The larger JAIN community is a great bunch of people. I would like to thank the following individuals from this community for various discussions we had: Steve Davis (Ulticom) and Matti Drissin (Ericsson Infotech) for equipment, Shmuel Kallner and Zygmunt Lozinski of IBM for discussions on both history and details of AIN deployment, Ravi Jain, Margaret Nilson, and Pualo of Telcordia for discussion at Supercomm and the July meeting, Colm Hayden and Aidan McGowan of APIoN.

Thanks go to the authors of several free software projects including: Linus Torvalds, Alan Cox and the rest of the Linux-kernel community, the OpenBSD and OpenSSH team, Brian Paul of Mesa, /.,lwn.net, and Richard Stallman. Without Richard's drive we wouldn't have any GNU, and the BSD source may not have been released. The lack of either of these would be a real pity.

Last but not least I'd like to thank my family for supporting me during this project and encouraging me, particularly when I didn't want to tidy the dissertation! I'd also like to thank both Dad and Nicola for proof reading.

Overview of the contents

Chapter 1 introduces the concept of Telecommunications services and notes the problem of service portability. The JAIN framework is introduced as providing a possible solution to the problem of service portability. The objectives of our research are also presented. Chapter 2 discusses the technologies present in the current wired-line telephone network. This includes an outline of the evolution of the wired-line telephone network, an explanation of the current day architecture named the AIN – Advanced Intelligent Network – and an overview of the protocol stack used in the AIN. The JAIN initiative is then discussed and is followed by a survey of related work. Chapter 3 introduces related areas of work in both telecommunications services and recent network architectures. Other Java-based telephony initiatives are presented and discussed. Chapter 4 illustrates architectures for two JAIN based services: Internet Call Waiting and Click-to-Dial. A summary of the work presented in this thesis and recommendations for areas of possible future work is finally presented in Chapter 5.

Chapter 1

Introduction

1.1 Telecommunication services

Telecommunications service providers (hereafter referred to as carriers) traditionally supply the necessary infrastructure to enable telephone calls to be made. As a customer typically has a choice between many possible carriers, carriers are required to differentiate themselves from competition if they are to maintain their existing customer base or expand it. A method used by carriers to attempt to achieve this differentiation is by offering *Telecommunications services or features.*¹ Once subscribed to a feature the customer receives functionality that is not delivered in a normal call. From the carrier's point of view, the offering of features also has the beneficial side effect of creating new revenue opportunities.

¹There is a distinction between a service and a feature. In both the ITU-T – International Telecommunications Union (ITU-T, 1992) – and Bellcore (Bellcore, 1991) standards, features are portions of services that the service subscriber can distinguish. Hence a service may contain several features. The terms *service* and *feature* are used interchangeably throughout the text

Some examples of services include:

- The voice mail service. This is a service where a caller is able to leave a message in the called party's *voice mailbox*.
- The 800-number service. This service allows people to dial a number toll-free. The called party is charged for the call.
- The calling card or alternative billing service. This service allows a subscriber to charge a call to a particular account regardless of the caller's physical location.
- Time based routing. This service allows a subscriber to have a call to a particular number redirected to another number based on the current time.

As many carriers provide services it is important for carriers that they are able to create and deploy services throughout their network quickly and cost effectively in order to maintain differentiation from competitors. This requirement has led to the introduction of a number of highly effective technologies. The network that provides call setup and termination is highly fault tolerant. At present, services can be rapidly created and deployed through the use of graphical representations of both components which may be pieced together to form services, and the network which the services are deployed on. Standards exist that ensure interoperability between equipment vendors.

1.1.1 JAIN

In spite of the many desirable aspects of current service creation techniques such as rapid creation, ease of deployment and inherent fault tolerance, they suffer from the serious drawback of non-portability. While interoperability standards exist ensuring communication between services executing on different vendors' equipment, a service that is created on vendor A's platform would have to be redeveloped to execute on vendor B's platform. This is due to the lack of standard programming interfaces. The JAIN program (Sun Microsystems, 1999a, 1999b) is aimed at addressing this problem and extending the service creation paradigm. JAIN is a set of open application programming interfaces (APIs) for the Java programming language (Gosling et al., 1999) that both include and extend the scope of traditional telecommunications service development. That APIs are open means that their specification is in the public domain. Futhermore, they are standard extensions to the Java platform. Standardization is important as any implementation of a JAIN interface must pass a compatibility test suite ensuring that it functions as anticipated. The Java platform allows software written in the Java programming language to run on any combination of operating system and hardware without modification to the software. Services can now be written in Java to use the JAIN APIs and execute on any vendor's platform which supports both the JAIN APIs and Java run-time environment.

Our work considers the above issue of service portability in some detail and studies the JAIN approach through the discussion of services that use the JAIN framework. It includes a number of objectives. These are as follows:

- A survey of work relating JAIN to existing work in the field of telecommunications.
- Identification of alternative frameworks or architectures that are similar to JAIN.

- Discussion of other Java based telephony initiatives.
- Illustration of JAIN's suitability for the development of portable services by building one or more services that demonstrate the value of a vendor and platform independent framework.

An understanding of the workings of services in the present day wiredline telephone network and telecommunications service development in general will help in appreciating the value of a JAIN based approach. This is therefore discussed in some detail in the following chapter.