

Investigating Standardisation Progress of Peer-to-Peer Session Initiation Protocol and Evaluating Existing Systems

CONVERGENCE.

RESEARCH

GROUP

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- Objectives
- SIP in summary
- P2PSIP
 - Distribution Model
 - Protocol Layering
 - DHT Choice
- P2PSIP Tested Systems
- Candidate systems for use
- Conclusion

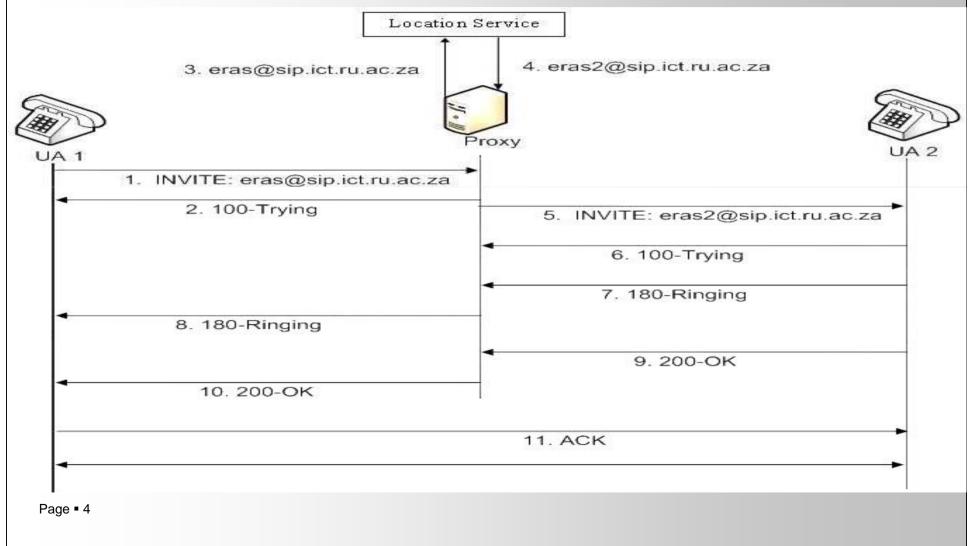
Key Issues for standardisation



- Investigate the standardisation progress of P2PSIP.
- Test P2PSIP implemented systems and identify which one is appropriate for further research use.

Session Initiation Protocol A signalling protocol used to establish, modify and tear down session over

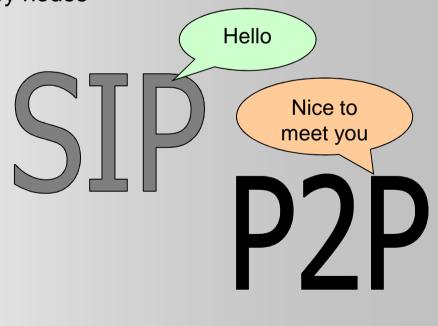
IP network.



- Based on client-server and consists of user agents, proxies and registrars
 - Proxies and registrars are only a mechanism to find an AOR
 - Proxies only route SIP messages (exercise almost no control)
 - User agents have usually more intelligence than SIP proxies
- Problems with client-server architectural model
 - Proxies and registrars create single point of failure
 - Limited scalability
 - High cost for maintenance

Overall suitability for emergency scenarios?

- P2P uses direct client-to-client communication.
- To build a P2P network that uses SIP, user agents need lookup and routing functionalities. That is, a DHT overlay needs to be built in the user agents to provide them with join, leave and lookup capabilities.
- SIP can allow this with little changes BUT there is a need to agree on:
 - Distribution Model: "Roles" played by nodes
 - DHT algorithms to use
 - SIP-using-P2P vs. P2P-over-SIP



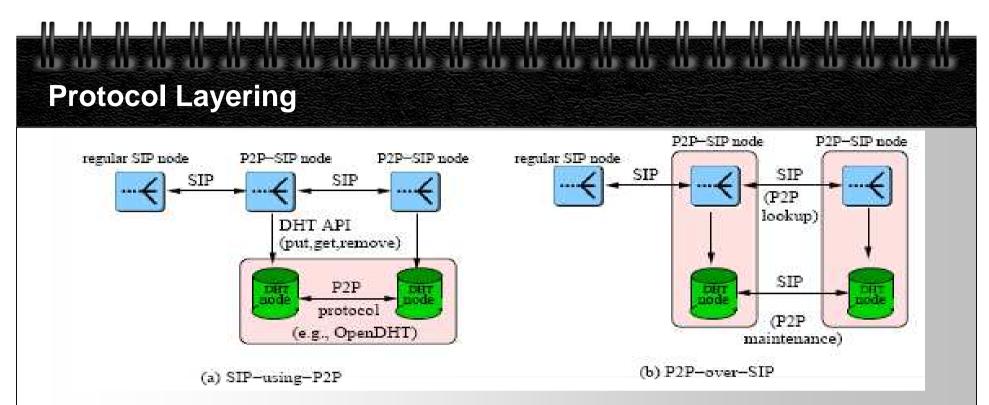
Distribution Model in P2PSIP

How nodes are distributed in order to carryout functions of central servers.

- Pure P2P model
 - All peers are equal
 - Any peer can carryout the functions of the removed servers

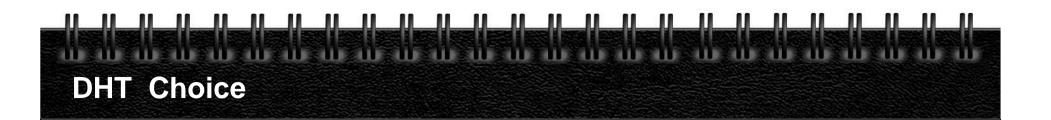
Partial P2P model

- P2PSIP Peers: It is a node participating in a P2PSIP overlay that provides storage and routing services to other nodes in the same P2PSIP overlay.
- P2PSIP Clients: It is a node participating in a P2PSIP overlay that does not store resources, run the distributed database algorithm, and is not involved in routing messages to other peers or clients.

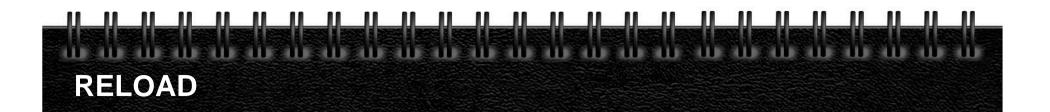


Architecture approach (from [Kundan PHD thesis])

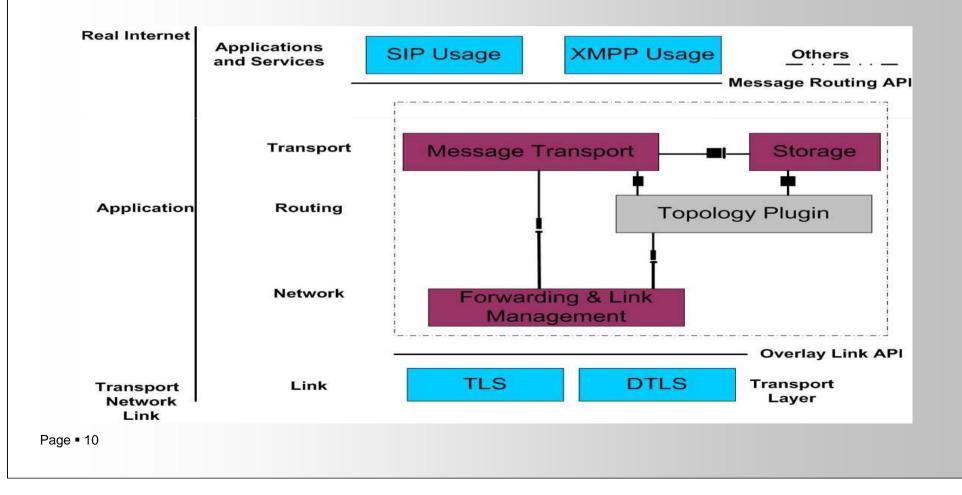
SIP layer for registering users, resource lookup, establishing session. a P2P layer for maintaining a distributed network. SIP messages are not used only for registering users, resource lookup, and establishing session but also for maintaining a P2P network. Overloading SIP causing high overhead, interoperable with other P2P applications

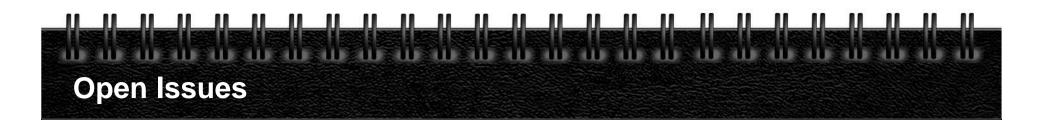


- Distributed Hash Tables (DHTs) is a class of decentralized distributed system that provides a lookup service.
- Commonly used DHTs:
 - Pastry
 - CAN
 - Bamboo
 - Chord
 - Kademlia
- Single vs. Multiple DHTs
 - Is it mandatory to implement DHT, or do we just specify one to use for testing?
 - Best approach



RELOAD is a P2P version of SIP. Adapted by the P2PSIP Working Group as its starting point for the primary P2PSIP protocol.





- The P2PSIP IETF is working on improving the RELOAD protocol
- Security
 - Connection level: Connections between peers are secured with TLS or DTLS.
 - Message Level: Each RELOAD message must be signed.
 - Object level: Stored objects must be signed by the storing peer.
- NAT ISSUES
- RELOAD support of Video on Demand

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P2PSIP Tested Systems

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	OverCord	39 Peers	SIP2P	OpenVoIP
Language	Java	Python	C++	C++
NAT support	No	Yes	No	Yes
Peers and Client	No	Yes	No	Yes
Complete	Yes	No	Yes	Yes
DHT	Bamboo and Open Chord	Kademlia	Kademlia	Kademlia, Bamboo and Chord
Hash Algorithm	None	SHA1	None	SHA1, MD1
Routing Methods	Recursive	Recursive	Recursiv e and Iterative	Recursive

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- Java implemented systems, developed at Rhodes University as a part of Master's thesis.
- Jain SIP Applet Phone

	Initialising Looking for installed plugins		
	2 plugin(s) loaded. Names are:		
### Finished overlay discovery and bootstrapping module ### ### Node instantiation ### Using plugin: chordplugin.ChordPlugin Starting: 146.231.123.76 as bootstrap peer Creating overlay: chord.ru.ac.za Booting parameters: 146.231.123.76 0 Running node on 146.231.123.76 on port 3730 [main] log4 configured with 'log4j.properties ****** Node options ***** 1. Insert record 2. Retrieve record 3. Remove record	chordplugin.ChordPlugin bambooplugin.BambooPlugin You can start an overlay with any of the following : > > 1. chordplugin.ChordPlugin > > 2. bambooplugin.BambooPlugin Your choice:		
4. Leave overlay Your value: [0] Page = 13			

Jain SIP P2P Applet Phone

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🛃 NIST Messenger	
Menu Settings Help	
Call:	(Online)
Buddies DTMF	
mtsietsi@chord.dyndns alfredo@chord.dyndns.o erasmus@chord.dyndn	org(offline)
ADD NEW RE	MOVE

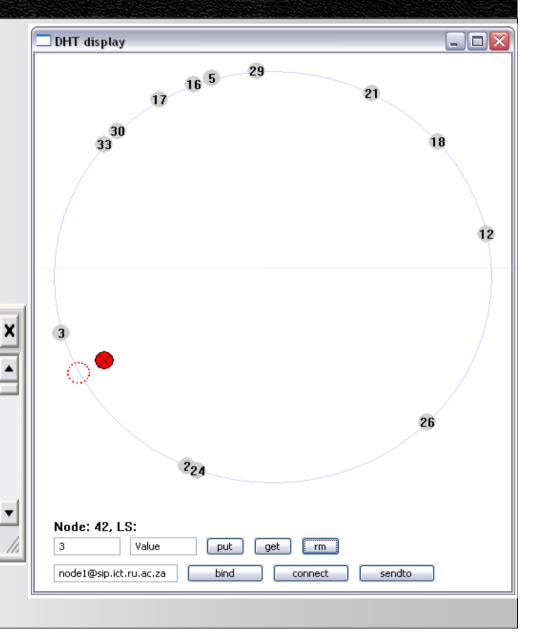
- SIP2P is P2PSIP proxy Published at source forge (<u>http://sourceforge.net/projects/sip2p/</u>)
- Based on early proposal candidate (Peer-to-Peer Protocol) for P2PSIP.
- Running on Linux machine
- Using X-Lite running on any machine.
- Not incorporated into a user agent, the ideal way is to incorporate SIP2P into a user agent.
- X-Lite

ccount	Voicemail	Topology Presenc	e Transport	Advanced
	Account name:	erasmus)	
	Protocol:	SIP		
User Deta	ails			
	💐 User ID:	ТуараЕ		
	Domain:	146.231.123.70		
	Password:			
	Display name:	Туара		
Auth	orization name:			
	ter with domain a bound via:	and receive calls		
) D () P		: (146.231.123.70		
0 P				
• P	roxy Address			

Candidate Systems for Use

- 39 Peers (<u>http://39peers.net/</u>)
 - Incomplete
 - Work like an adaptor running locally
 - Implemented user agent class to allow user agents to route calls
 - Support NAT using STUN, TURN

C:\WINDOWS\system32\cmd.exe - python ap... _ C:\p2p-sip\src>python app\dhtgui.py put Key, Value connect node1@sip.ict.ru.ac.za sendto node1@sip.ict.ru.ac.za Value connect erasmus@sip.ict.ru.ac.za sendto erasmus@sip.ict.ru.ac.za Value

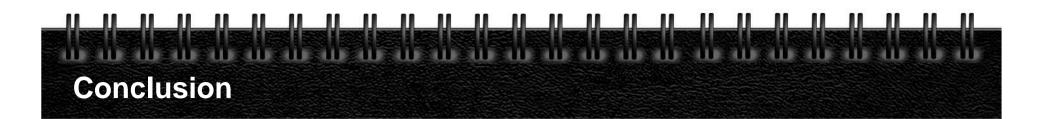


Candidate Systems for use

- OpenVoIP (Columbia University)
 - Borrowed code to create our own P2P overlay.
 - Created a bootstrap server running on a local machine.
 - Users can join as client or as peers
 - Our system does not support NAT
 - Users use OpenWengo-P2P phone

🖾 C:\WINDOWS\system32\cmd.exe - p2p.exe bootconfig.txt logfile	- 🗆 🗙	
D:>>cd p2pp	^]
D:\p2pp>p2p.exe bootconfig.txt logfile 3 Algorithm is KademliaBootstrap server >>Message received from UID Sacha IP: 146.231.123.80 port: 7080 packet IP: 146.231.123.80 packet port: 7080 Generating a SHA1 ID 4a3c8e35b98074a4cd74fdcf44a5a73834e7dca8 Message received from UID Sacha IP: 146.231.123.80 port: 7080 packet IP: 146.231.123.80 packet port: 7080 Generating a SHA1 ID 263278d99e505bd11b056ca815c9448259269100 Message received from UID Mukole IP: 146.231.123.58 port: 7080 packet IP: 146.231.123.58 packet port: 7080		1
Generating a SHA1 ID 5e124f0c2c33f44add207ff08ee0521125d3443f		
		2

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- P2PSIP has been put in practice in Industry (e.g. Damaka) and Academia (Columbia University).
- There is lack of implementations.
- Project under development and lack of technical documentation
- Lot of bugs
- Security and NAT issues need to be addressed.



Thank you for you attention

• Questions?