



Fraunhofer Institute for Open Communication Systems

SIP Operation in 2003

Iptel.org – builders of SER Jiri Kuthan, Founder

http://www.iptel.org/

International SIP, January 2004, Paris

About iptel.org

- iptel.org is a SIP know-how and deployment organization -- it created world's most unique open-source SIP server with premium service creation flexibility and performance. The server has been powering iptel.org's public services as well as services of iptel's customers.
- iptel.org spun off from Germany's national research labs, Fraunhofer, home of MP3 and very first implementations ever of mobile IP and IPv6 applications – see www.fokus.fhg.de.
- iptel.org provides software, consultancy and technical support to both operators and vendors in the SIP area.

2003 SIP Landscape

- Construction is Over Operation Began, Perfection on Agenda
 - Pre 2003

2003



Deployment Roadblocks Vanished

- Affordability:
 - 2003 the first SIP telephones bellow \$100 marketed
 - Scalable network solutions available (picture shows installations of SER, <u>www.iptel.org/ser/</u> in early 2003)



... Deployment Roadblocks Vanished

- Solutions for technological headaches matured:
 - Number #1: NATs (in average 20% of population behind hard-to-traverse symmetric NATs) – technology to traverse NATs exists and is deployed
 - Interoperability proven: variety of compliant devices
 - Application building matures and replaces naïve or monstrous API concepts.
 - Scalability: SER/discount PC offer capacity to power Bay Area.
 - QoS Issues: Where Are Thou?





- See pictures for example of packet loss measurements in Scandinavia
- Modern end-devices can cope with QoS distortions



- ITSPs and ISPs offering IP telephony on top of IP access; telephone line no longer needed.
- Business model changes seat order: initial investment barrier small enough to be overcome by a variety of competitors.



Case #1: Affordable ITSP w/PSTN Connectivity

 addeline.com: Texas-based ITSP with focus on affordable telephony; cost-effective deployment powered by Linux PCs; free basic service, subscribers may receive PSTN telephone number in any of 17 areas, calls follow their SIP devices; international minute rates between \$0.060 (Buenos Aires, Austria, Australia, Belgium, China, ...) and \$1.807 (Thuraya, satellite); monthly fee \$12.95 includes 1000 local and 200 continental minutes, additional minutes at \$0.03 per minute.





- VozTelecom: Spain-located SIP-based ISP
- Premises: value is in novel applications, innovation of telecom technology blocked by infrastructure cost and closed service model -> move to VoIP!
- Answer: Web-SIP Network Architecture which opens up creation of services to third parties





Case #3: Replace Phone Line with DSL/SIP

- Situation: Deregulation of the Norwegian telecom market, number portability obligation for incumbent, termination at regulated cost based prices: consumer can use copper for ADSL <u>without</u> having PSTN subscription
- Competative price structure (next slide)

Telio.no – Advantage in Cost Structure



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Case #4: Campus Networks

- Premises: cost-effective migration from Centrex to integrated IP services
- Yale University Scope:
 - Altogether about 90,000 calls
 - Assuming three minute hold, about ¼ million minutes
 - 50-100 phones
 - Serving phones located nationally
 - Phones = Cisco, Pingtel, Mitel and Grandstream
 - Softclients = Messenger, SIPc, Session, Xten, etc.
 - About 15,000 aliases used in SER that make every telephone (IP and circuit switched) at Yale reachable by URL dialing.

Conquest: World-Wide Reach

- The service is replacing classic static fixed line telephony services by flexible and mobile communication solutions
- Accounts are available worldwide and come with full Unified Messaging Features (Mailboxes, Call Forwarding, ...). Customers can create and modify all account details via web-admin.
- Customers can get unbundled by either assigning a geographic number to them (+43 720 ...) or by porting their existing number. Classic landline services provided by PTT can than be cancelled!

Bringing SIP to Perfection

- Integration: like many other markets, VoIP is moving from commodities (solutions, hardware, service components, ...) to where the hard part is: services (integration, accomplishing availability, service building, etc.). Problem: integrators don't have routine experience yet and SIP devices not yet very PnP.
- Architectural Sanity
 - Stay ready for new generation of networks: support IPv6
 - Note: iptel.org's site offers SIP/IPv6
 - Keep the design managable

More to Tune

- Security. There is no widely deployed interdomain trust model yet. Problems:
 - Spam: Do really European subscribers wish to receive calls from US telemarketers at 3 AM? ("Want to make some money", "Add 3+ Inches Today", "Don't gain the winter weight")
 - Identity and Fraud Prevention: If a user from a domain terminates to PSTN via another domain, how does the terminating domain learns a trustworthy Caller-ID to propagate to PSTN and charge to?
 - Solution: use web-proven TLS to establish Internet-wide trust.
- Keep interoperability manageable:
 - SIP Forum Testing Group: focus on interoperability issues known to cause troubles in field and interoperability events
 - ETSI: focus on formal verification (TTCN)

Final Observations

- SIP began penetrating in 2003 most deployments aim at cost-effectiveness which is easy to achieve through very low introductory cost barrier.
- Sign of raise: The technological market begins to mature from commodities to services.
- Still on the agenda: making SIP easy-to-integrate, plug-and-play, establishing security and trust interdomain models.

Acknowledgments and Disclaimers

- Acknowledgments: useful information has been provided to presenter by courtesy of iptel.org, telio, voztele, and addaline.
- Statements made in the presentation may or may not be shared by the respective companies.
- Disclaimer: The presentation has been focused on technology: remember politicians still keep power to spoil what technologists have done.

Information Resources

- Email: jiri@iptel.org
- IP Telephony Information: http://www.iptel.org/info/
- SIP Services: <u>http://www.iptel.org/user/</u>
- SIP Express Router: http://www.iptel.org/ser/