

# • OpenSIPS, what you can do.

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## **Cloud Services for Business**

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- What's SIP?
- Proxy Servers
- OpenSIPS
- Question?









• SIP is a Voice over IP Protocol





- H.323 (D-Ch/ISDN over IP)
  - ITU-T (Telecommunication Standardization Sector)
- SIP (Text based similar to HTTP)
  - IETF (Internet Engineering Task Force):
- SIP wins:
  - simpler (Text based)
  - more flexible (presence, IM, app sharing, special features)
  - NAT/Network traversal capability
- Media: RTP/RTCP (over UDP)





History











### Started in 1892 - First phone call







- Telephones needed cables (still do)
- Lots of Cables
- The art of cabling
- Connecting Terminals, Patch Panels, etc.

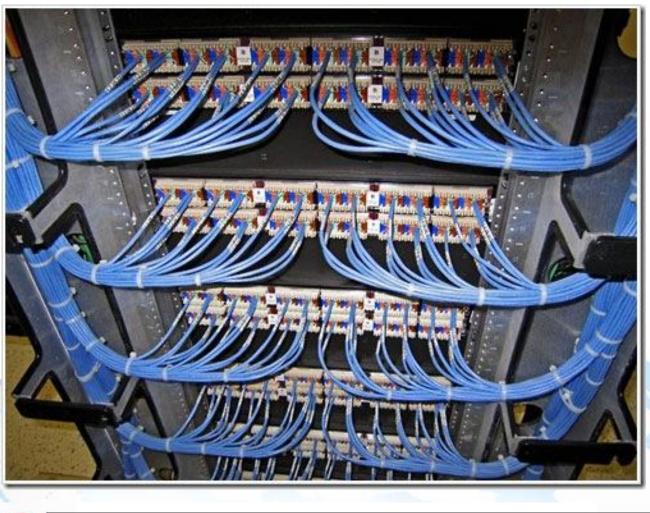








# Distribution Panels, Connecting Terminals, Patch Panels, Connection Blocks, etc.







- IP: Firewalls, Routers, NATs
- ISDN Gateways: Digital and Analog
- Electricity: Distribution Panels
- Water Supply Distribution System
- Airport transits



When it comes to VoIP, you need OpenSIPS





- Proxy Server
- SBC
- Back to back user agent
- SIP Router





#### Advantages of using a proxy server

- Improved performance
- Security and Authentication
- Flexibility and Customization
- NAT
- Redundancy and High Availability (failure detection and re-routing)
- Distribution
- Routing, Rules and Priorities





## Advantages of using a proxy server (Cont'd)

- Load balancing
- SIP Signaling, Codec and Digit Manipulation
- Scripting
- Resource Allocation
- Rate Limiting
- Media Proxy (or Direct Media Path)





- No Cost, no licensing fees
- Flexibility, Customization and Integration
- More independence (not depending on a single company)
- Community
  - Wider support
  - Continued enhancements
  - Tools
- In line with future of telecom





- Call Routing (Rules, Priorities, LCR & Reporting)
- NAT Traversal
- Security
- Registrar Server
- SIP Signaling, Codec and Digit Manipulation
- Load Balancing
- Failure detection and re-routing
- Scripting
- Database
- Management Interface





- Performance Over 1000 calls per second
- Distributed architecture
- Over 120 Modules
- www.opensips.org
- Learning Curve
- Mailing Lists, Forums, OpenSIPS Summits, etc.





- SIP Clients related modules:
  - Registrar Server
  - Authentication
  - Presence Server
  - Instant messaging, Jabber, Json, SMS
  - NAT Traversal
  - Back to Back User Agent (topology hiding)
  - Database (mysql, oracle, odbc, postgres)





- Trunks and Routing
  - Dynamic Routing (Prefix base rules, priority, time, distribution, drain mode, failure detection and rerouting)
  - Load Balancer (Dispatcher)
  - Dial Plan
  - SIP MSG OPS
  - NAT Traversal
  - TLS
  - XMPP (SIP to XMPP Gateway)





- Utilities
  - Rate Limit
  - Statistics
  - SNMP interface
  - SIP Trace
  - SIP Capture
  - Pike Flood detector
  - Perl
  - Accounting
  - Call Control (PrePaid application module)









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