

Architecting TheSIP Network

Today's Topics

- Infrastructure
- Security
- Interoperability
- Quality and Reliability



IP-PBX

ITSP

PSTN Gateway

INFRASTRUCTURE

WHAT'S IN YOUR NETWORK?

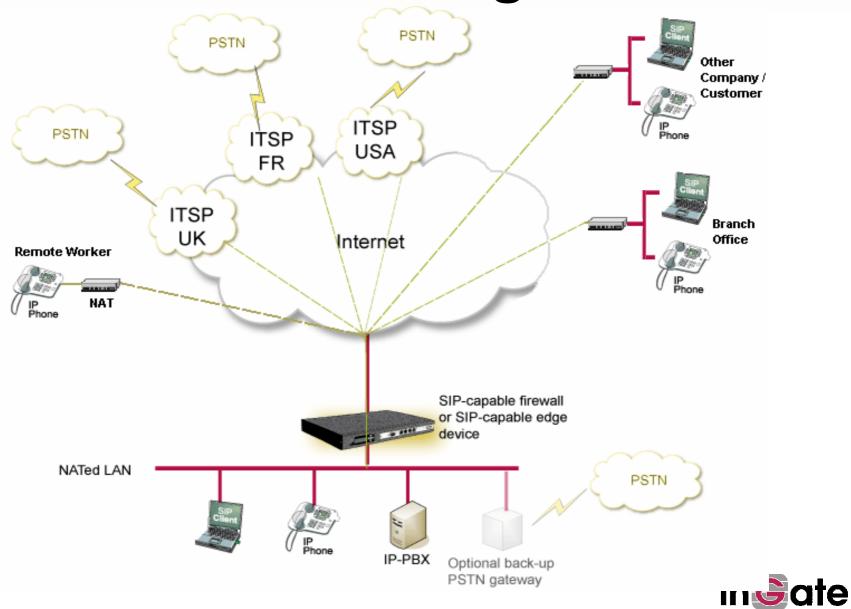
SIP Trunking

SIP-Aware Firewall

Remote Phones



Network Diagram



VLANs?

- Limiting No Rich Communication
- No Soft Clients

Single Network!

- Realizing Full Converged Communications
- QoS

Bandwidth Requirements

- LAN (Data, Voice)
- WAN (Data, Voice)
- Codec (G.711, G.729, others)



- Power
 - UPS
 - PoE
- Monitoring



IP-PBX Component

- Hybrid IP-PBX
- IP-PBX with SIP front end
- SIP Based IP-PBX

Things to Consider???

- Needed Functionality
- Connect Multi-Vendor Endpoints
- Manageability
- Integration with other IP-based Applications
- Remote Workers / Branch Offices
- Anonymous VoIP Callers (Other Systems)



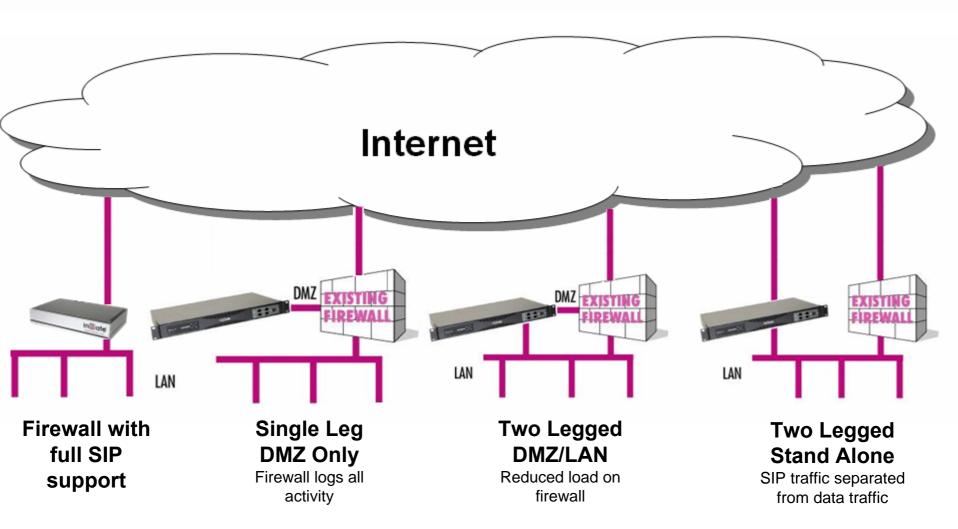
- Enterprise Edge Component
 - FW
 - NAT
 - SIP Aware (SIP Proxy Based)

What can it do for me???

- Security
- Mediation between IP-PBX and ITSP
- Reliability / Survivability Features
- Enable Remote Workers
- Branch Office Interconnect



Deployment Flexibility





- Service Provider Component (ITSP)
 - Facilities-based
 - Facilities-less
 - ISP w/ ITSP services

What can it do for me???

- PSTN Connectivity
- Local Breakout



Security

Threats

- Brute Force
- Denial of Service (DoS & DDos)
- Malformed Message attacks
- Buffer Overflow Attacks
- Session Hijacking & Redirection
- Voice Spam
- Packet level intrusion
- Vmail bombing
- SIP Spam
- VoIP Security Alliance (VoIP-SA)



Security

- Stable Platform
- SIP Signaling
 - SIP Parser
 - Optional Loop Detection
 - Authentication (DIGEST)
 - Encryption (TLS or MTLS)
 - Filtering (ex. Content type, method, To/From Header, black/white listing, etc.)



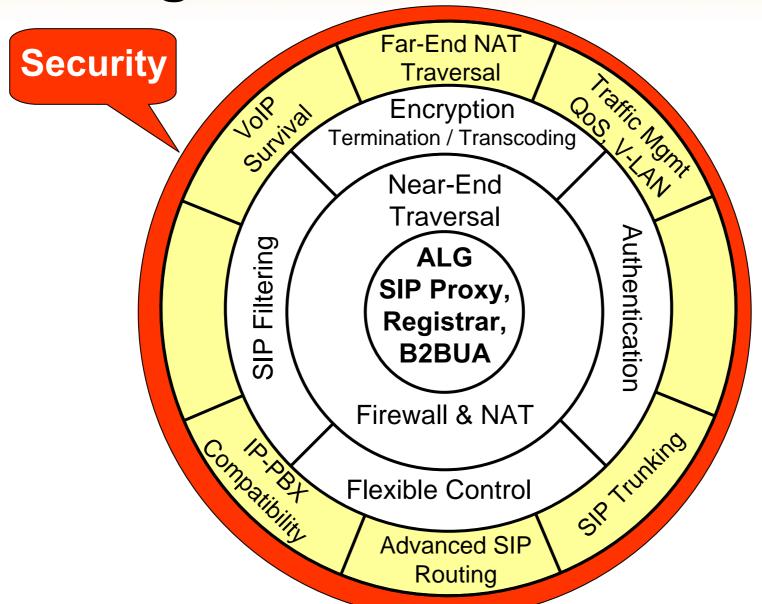
Security

Controlling Media

- Dynamic port control
- More secure than STUN/TURN/ICE
- Only accept media from endpoint that receives media (prevents media injection)
- Media Encryption
 - SRTP



Edge – Needed Feature Set





Interoperability

SIP Standards

- Defined by multiple IETF Standards
- Different vendors focus on different specs.
- Testing Important
- **SIPconnect** Created by SIP Forum as a set of best practices for interfacing PBX with ITSP.
 - DNS
 - Signaling Security
 - FW/NAT Traversal
 - PSTN & SIP Addressing

- Quality of Service
- Handling of Media
- Authentication
- Accounting



Interoperability

Mediation

- IP-PBX to ITSP
 - Varied Authentication Requirements
 - Least Cost Routing
 - Call Transfer
 - Conferencing
- Remote worker (behind SIP-unaware NAT)
- ENUM



Quality & Reliability

Prioritization of Voice Traffic

- Services (protocol & port)
- Packet size
- SIP Traffic
- IP address and segments
- > Inbound and Outbound
- > Dynamic bandwidth allocation
- Setting of Type of Service (TOS) and/or Diffserv bits



Quality & Reliability

- Call Admission Control
 - Emergency Calls
- Other Voice Quality Factors
 - PCs/Laptops
 - Fragmented disks
 - Poorly managed software installations
 - Sound card not optimized for voice
 - QoS on LAN
- MPLS
 - Voice only?



Quality & Reliability

- Reliability of SIP Trunks
 - UPS a must
 - TDM/PSTN gateway capacity as a backup?
- Failover to Secondary ISP
- Failover to Secondary ITSP
- Failover to Secondary Edge Equipment
 - Registrations and calls lost
 - Registrations maintained
 - Registrations and calls maintained





Bringing SIP to the Enterprise

Shane Cleckler
Mgr Systems Engineering

Mail & SIP: shane@ingate.com

Direct: 972-678-0464