

# **SIP:** Connecting the Distributed Workforce

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ITEXPO East  
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# today you'll learn about...

- ▶ sip, trunking, and ucc
- ▶ the evolved PBX architecture
- ▶ distributed worker case study
  - centralized SIP trunking, multi-site MPLS with QoS, collocation (“private cloud”), remote SIP devices, unified communications
- ▶ five tips for designing a distributed voip/uc infrastructure

## the mobile infrastructure experts

Mobile infrastructure includes the systems and tools that are used to **secure** corporate assets, remotely **manage** devices, and **enable** applications for workers who spend significant time **away** from the office.

## mobile device management

offline data sync

## software distribution

content management

asset tracking

configuration/pim management

remote control

data encryption

## mobile access

## mobile development

# customers

unwired  
revolution



GroupHealth



E.&J. Gallo Winery



WILLIAMS-SONOMA



TRADER JOE'S



OTIS



AMGEN

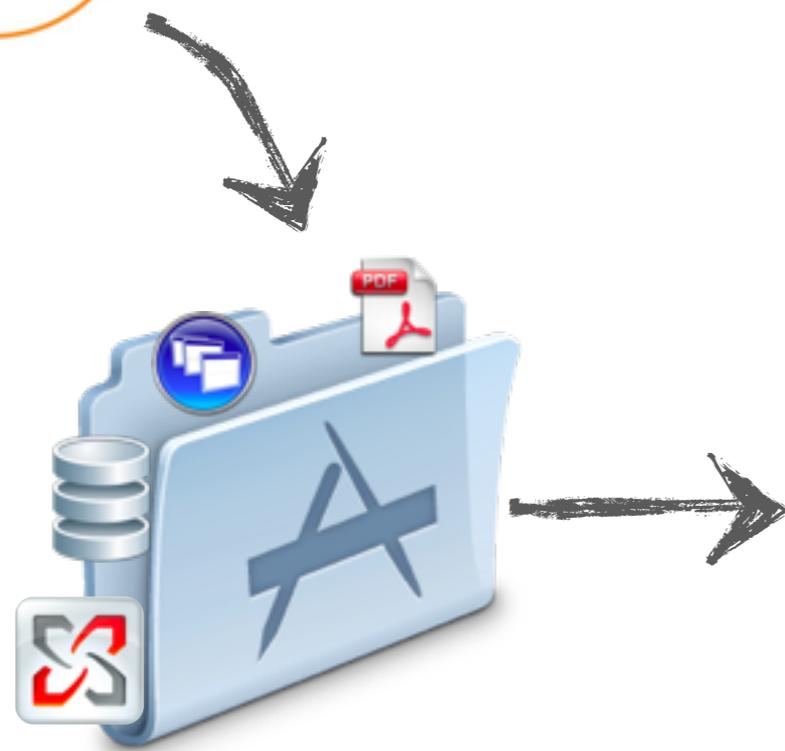


HOT TOPIC

AVISTA



# mobilizing uc



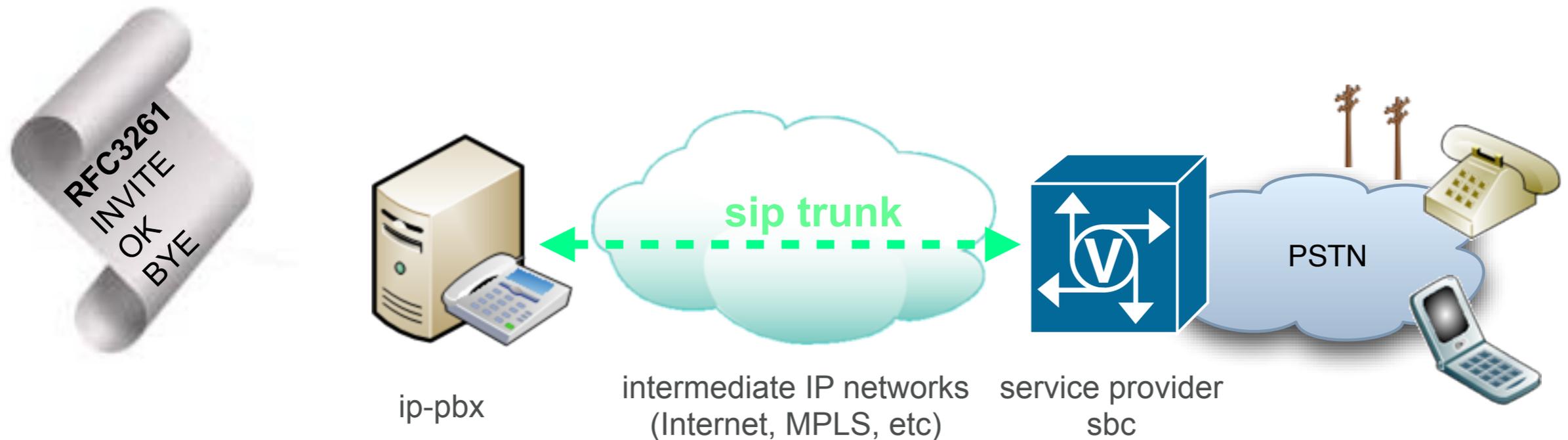
enterprise mobility platform



# sip, trunking and ucc

# sip + sip trunking

- ▶ session initiation protocol: standard defined by RFCs
- ▶ an IP-based “handshake” to set up, destroy, or modify a communications session between two systems
  - trunking: to the PSTN (public switched telephone network)
- ▶ connects using any IP network, including internet



# sip is here to stay

- ▶ adoption and large investments by key industry players



- ▶ open standard
  - promotes innovation and interoperability

- ▶ built to support multi-modal communications



voice



video



messaging



presence

▶ unified communications and collaboration

– real-time software applications that facilitate multi-modal communications and collaboration between two or more people



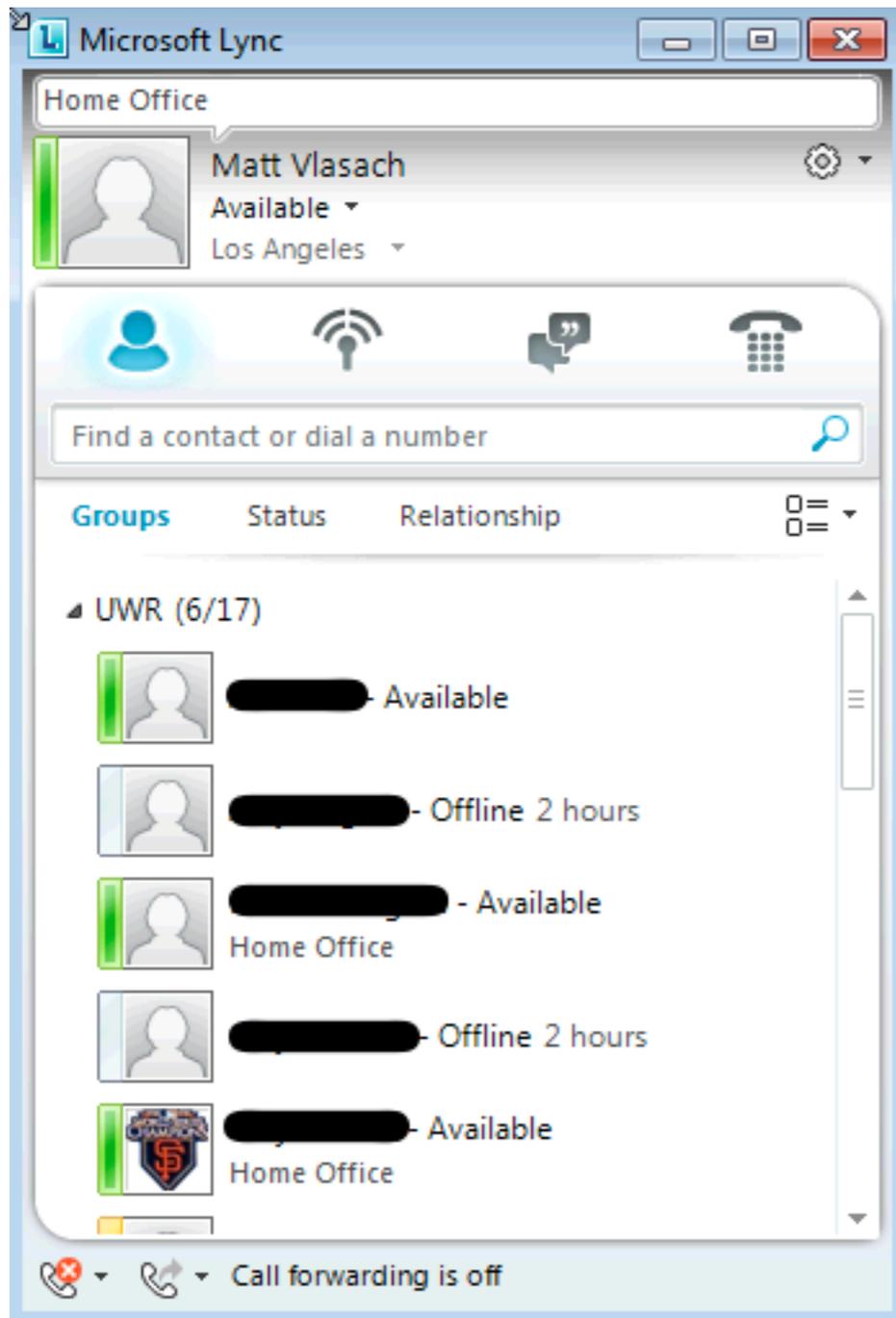
source: <http://www.unwiredrevolution.com>

- ▶ all vendors have different strengths/weaknesses
  - different backgrounds, priorities, dev cycles
  - no single vendor makes the silver bullet uc solution
  - perpetuated by continuous evolution of ucc tech

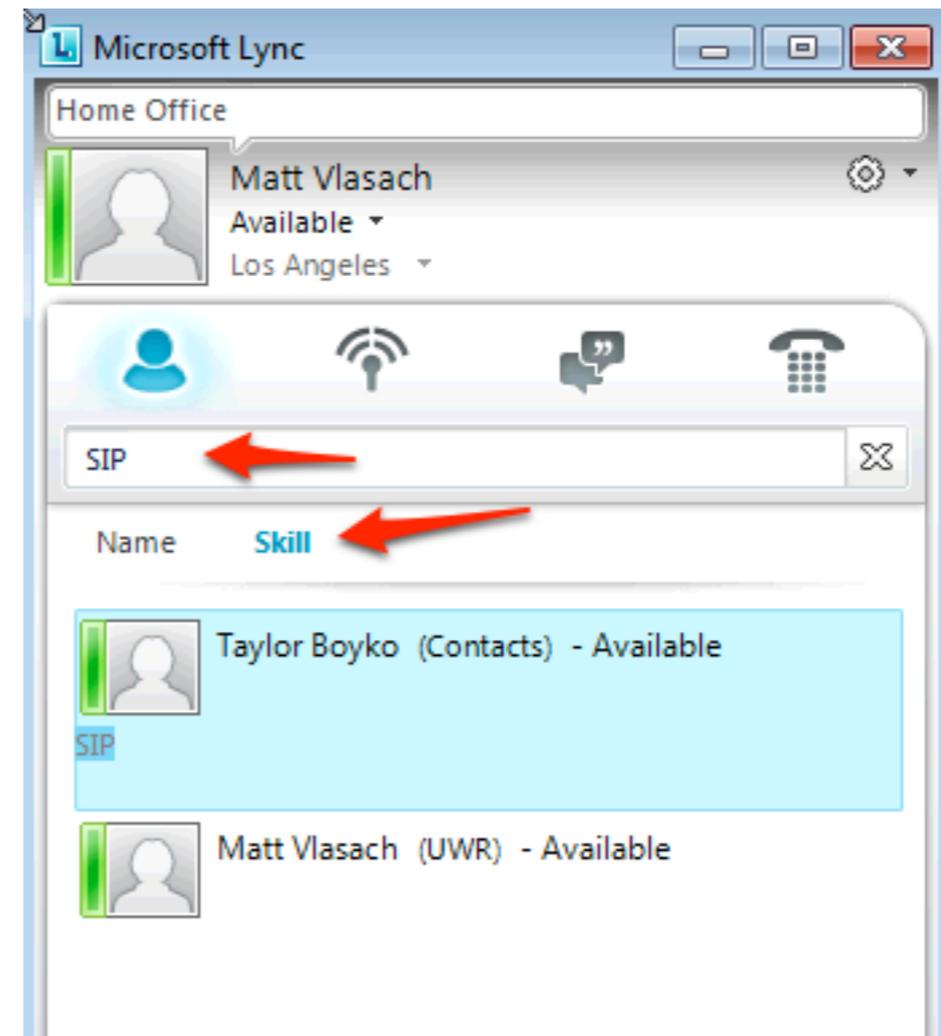
	vendor 1	vendor 2	vendor 3
telephony			
im + presence			
conferencing			
desktop video			

- ▶ new version significantly improved over OCS 2007 R2
  - simpler end-user experience, single client app
  - sharepoint and office applications integration
  - improved security certificate mechanisms
  - download-less web conference participation (silverlight)
  - dramatically improved mac support
    - voice, video, screen sharing
  - mobile client (iOS + WP7) and improved mac version “coming in 2011”

# lync 2010



main lync user interface



sharepoint skill-based searches

# lync 2010

Untitled - Meeting

File Meeting Insert Format Text Review

Delete Calendar Appointment Scheduling Assistant Join Online Meeting Meeting Options Cancel Invitation Attendees Options Recurrence Time Zones Room Finder Categorize Tags Zoom

Invitations have not been sent for this meeting.

From: MV Unwired

To: [Redacted]

Subject: Discuss Project

Location: Online Meeting

Start time: Mon 2/14/2011 12:00 AM

End time: Mon 2/14/2011 12:30 AM

[Join online meeting](https://meet.unwiredrev.com/mvlasach/BQD3SC80)

Join by Phone  
+1 (602) 714-XXXX

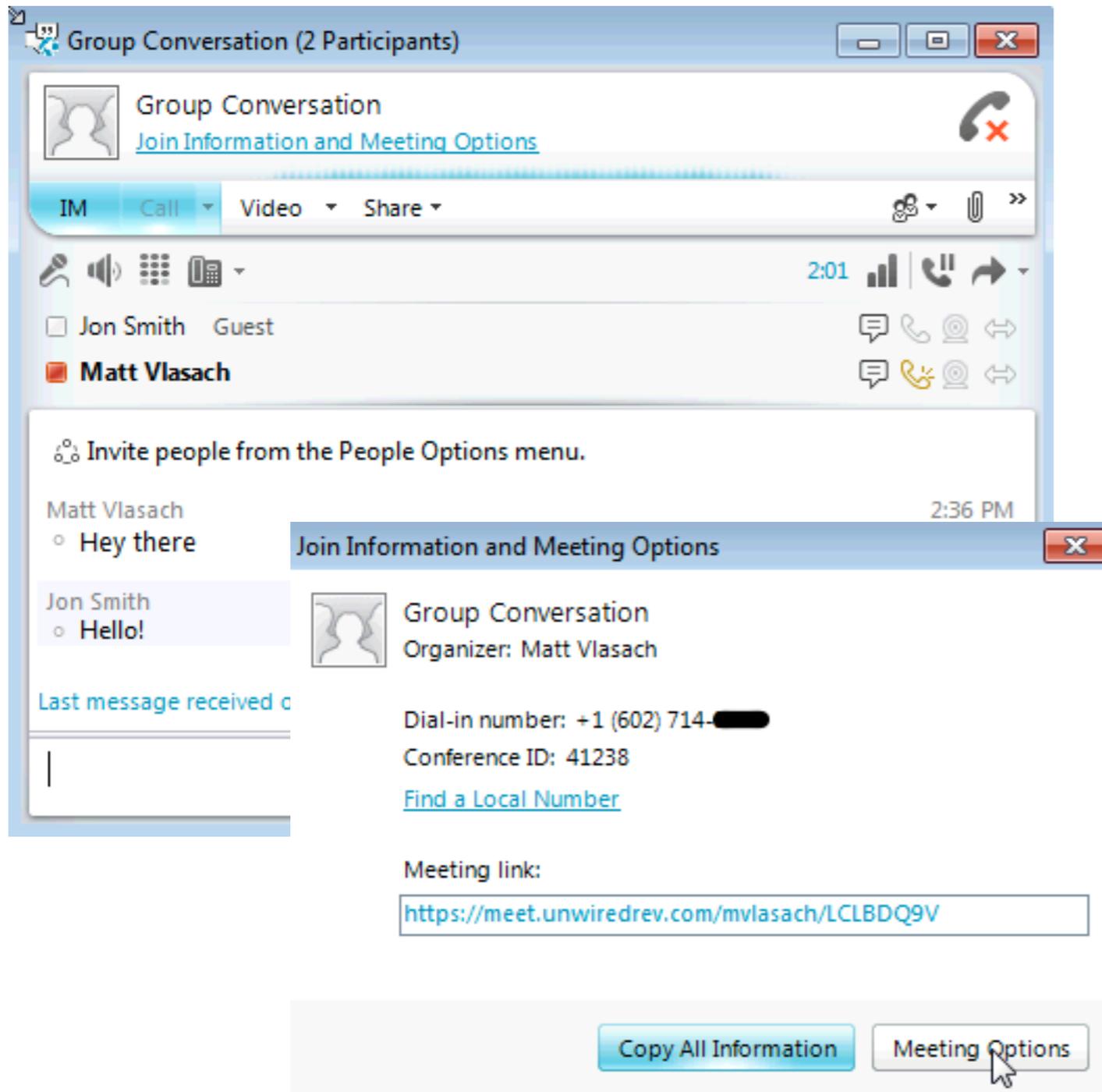
Conference ID: 58905

In Shared Folder: Calendar

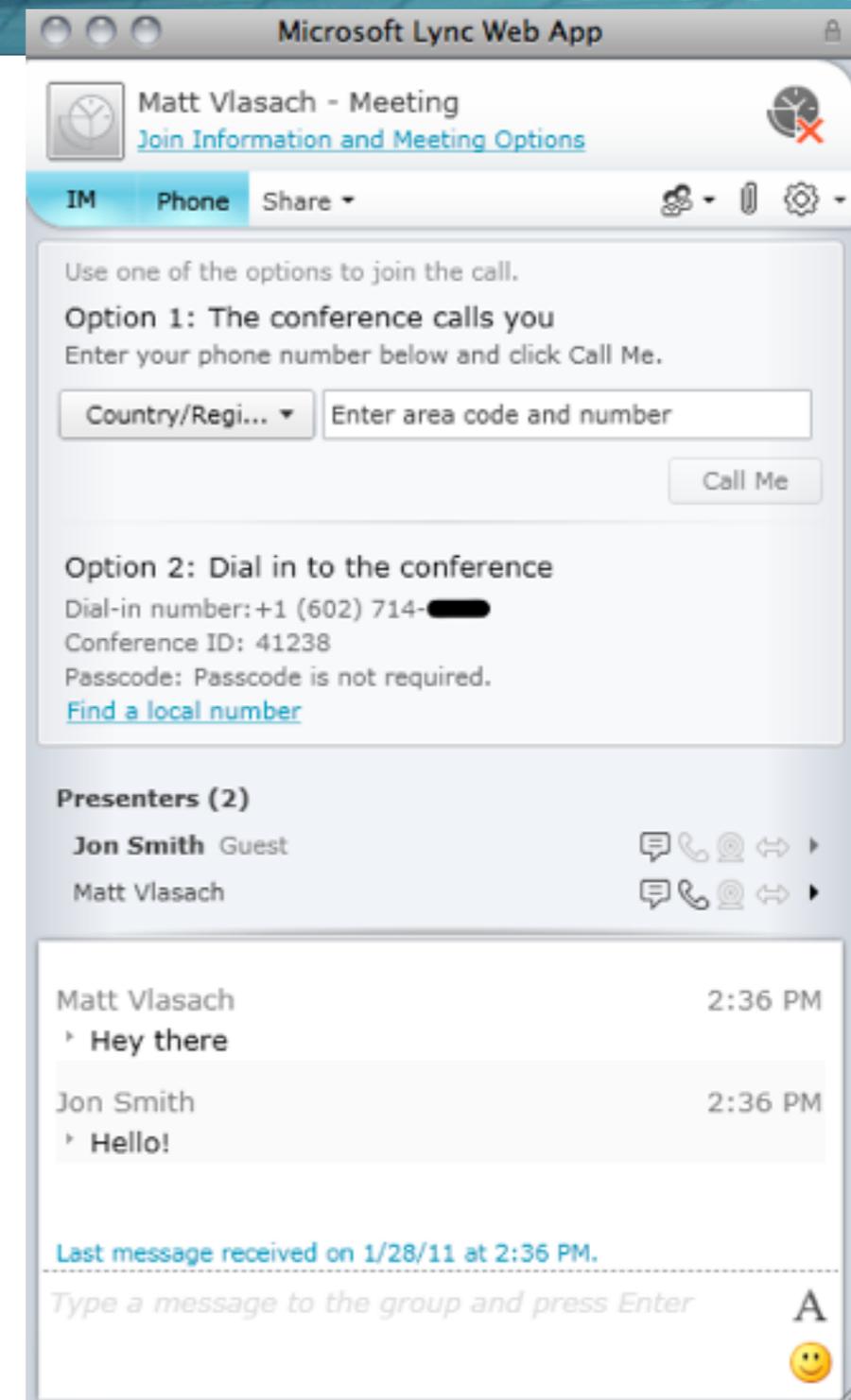
Matt Vlasach

outlook integration, meeting scheduling

# lync 2010

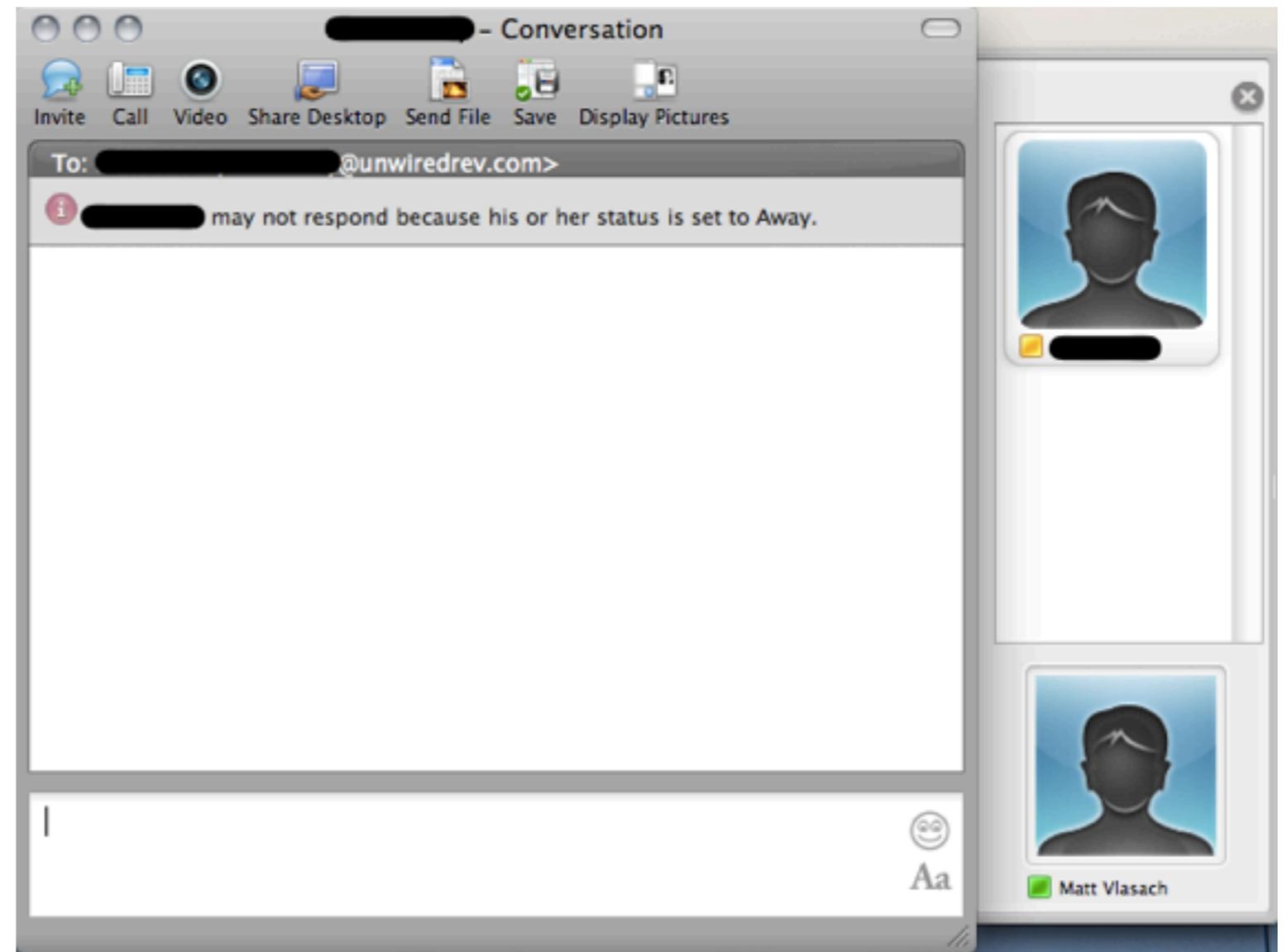
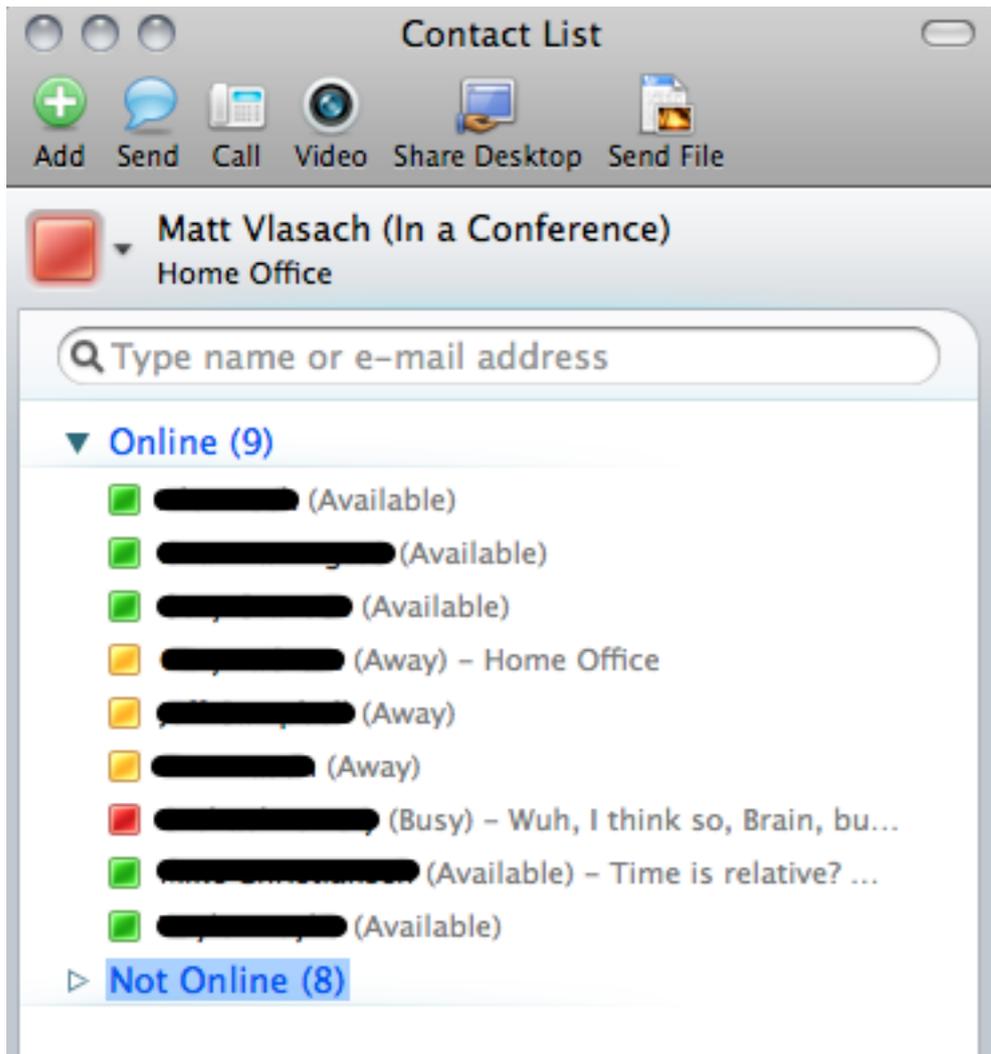


lync “conversation” window



“lync web app” silverlight viewer

# lync 2010

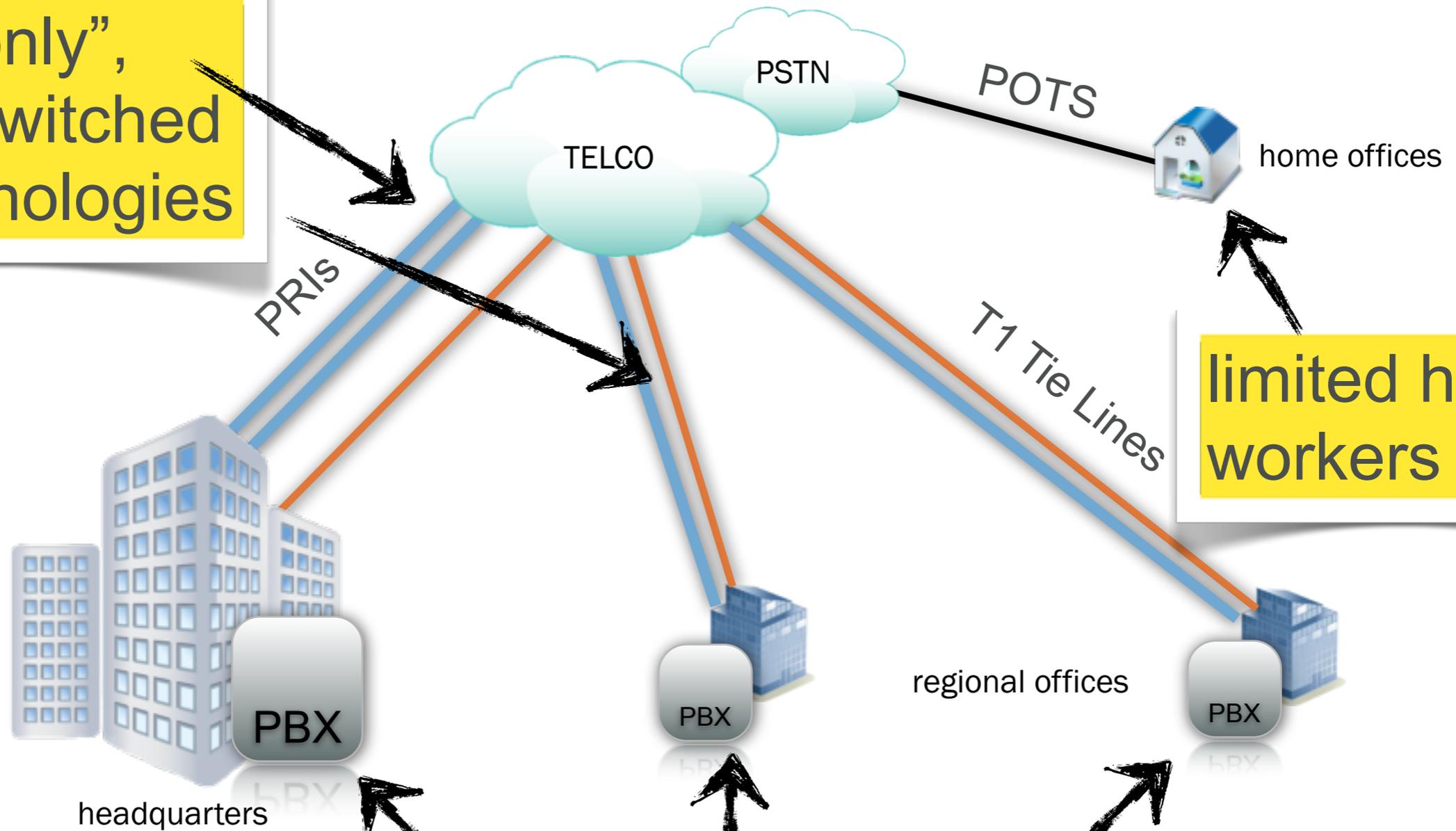


mac communicator

# the **evolved** pbx architecture

# legacy architecture

“voice-only”,  
circuit-switched  
\$\$ technologies



limited home  
workers

PBXes everywhere

# a decade of evolution

## element

## trend

broadband + mobile

faster, cheaper,  
more reliable, 'ubiquitous'

convergence

MPLS installations,  
multi-modal comm. with QoS

recession

cost cutting, real-estate  
reduction, staff cuts

cloud

public SaaS adoption,  
private datacenter

outsourcing

increasing domestic and  
global reach

highly  
distributed &  
collaborative  
workforce

# elements of the evolved pbx



- ✓ support for multi-modal communications
- ✓ utilizes IP network investments (eg MPLS)
- ✓ highly reliable, no single point of failure
- ✓ centralized management
- ✓ support for mobile and home workers
- ✓ user-friendly and business process integrated
- ✓ cost effective management and support

# the evolved pbx architecture

centralized,  
highly-available,  
server-based  
core



IP-PBX +  
UCC +  
Data Servers

datacenter

secure  
access from  
anywhere

scalable  
network access

PSTN

Service Provider

Internet

3G/4G



mobile workers



home offices

home broadband

QoS IP Circuits

regional offices



headquarters

# all connected by sip!



**case study**  
partner engineering and science, inc.

# about partner esi

## **PARTNER** Engineering and Science, Inc.



- ▶ nationwide environmental and building assessment company
- ▶ 6 company offices, 115+ endpoints
- ▶ 50 (~43%) workforce remote / mobile
- ▶ mostly knowledge workers
  - on-site data collection
  - highly collaborative
  - report generation
- ▶ uses microsoft office suite as primary productivity software

## im + presence

audio conferencing

on-net dialing

application/screen sharing

*it infrastructure requirements*

centralized trunking

“find me”

*user applications and services*

high availability

scaleable

fixed-remote voip

# solution components

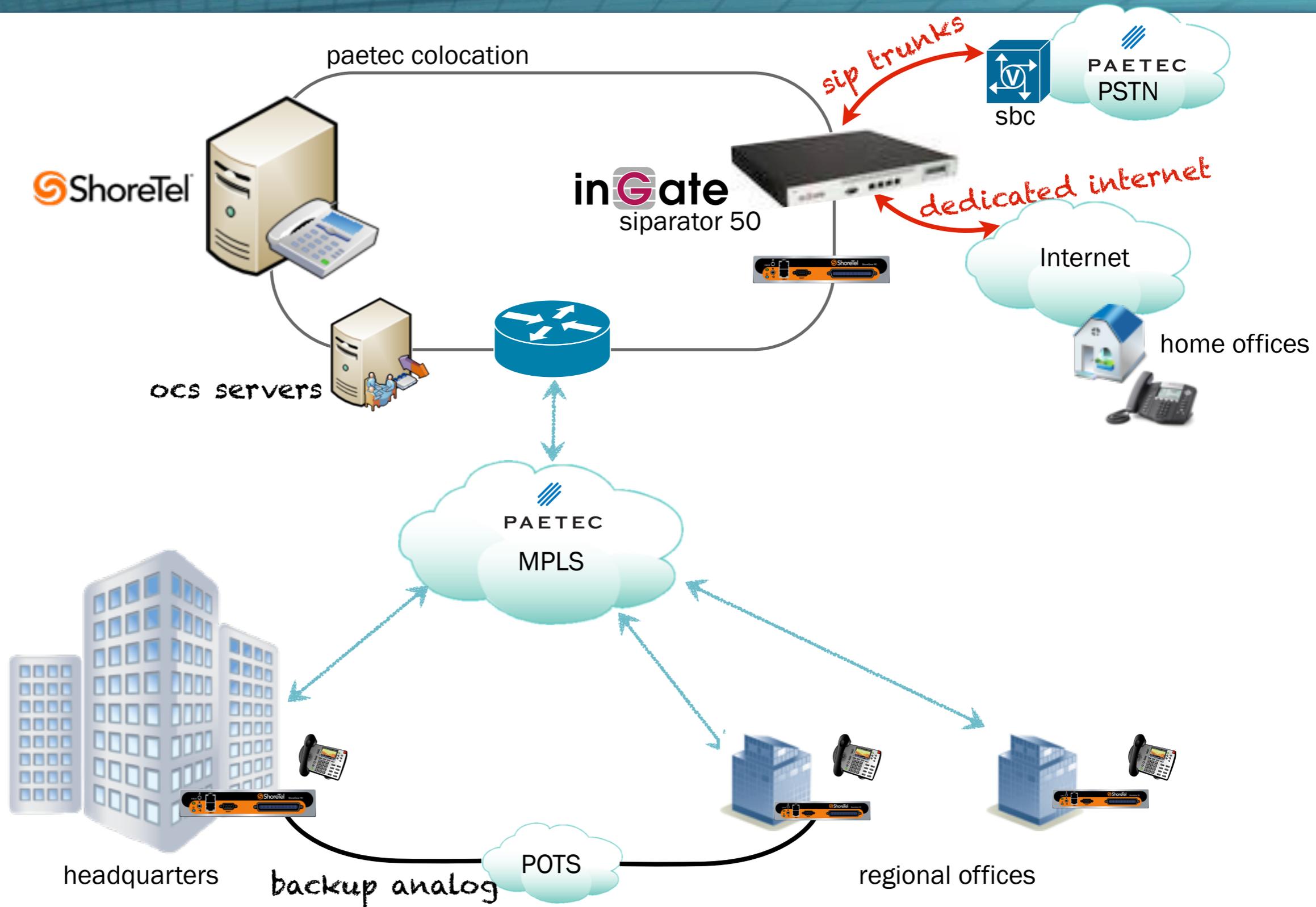
unwired  
revolution



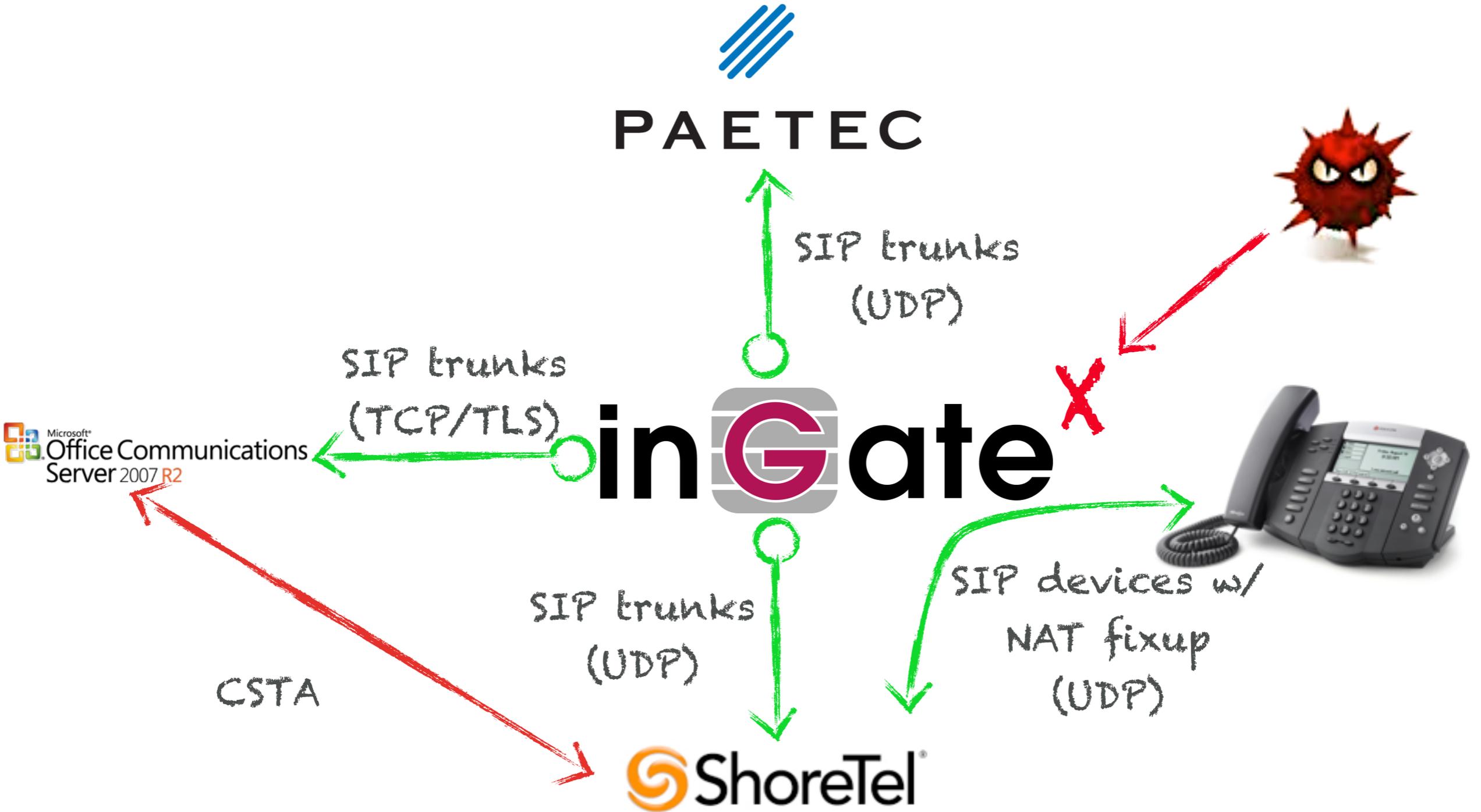
on-net SIP trunking	voip telephony + unified messaging	sip trunking interoperability	im + presence	SIP-compliant endpoints
mpls network with qos	find me / mobility support	remote SIP device (NAT fixup)	office integration with click-to-call	centralized provisioning support
colocation with ethernet handoffs	high availability via distributed architecture	SIP + IP firewall / edge	internal audio / web conferencing	hd voice (g722)
direct internet access	centralized management	transport transcoding (TCP<->UDP)	application and screen sharing	background noise suppression

# solution design

unwired  
revolution



# “the glue”



improved ad-hoc collaboration

scaleable and reliable platform

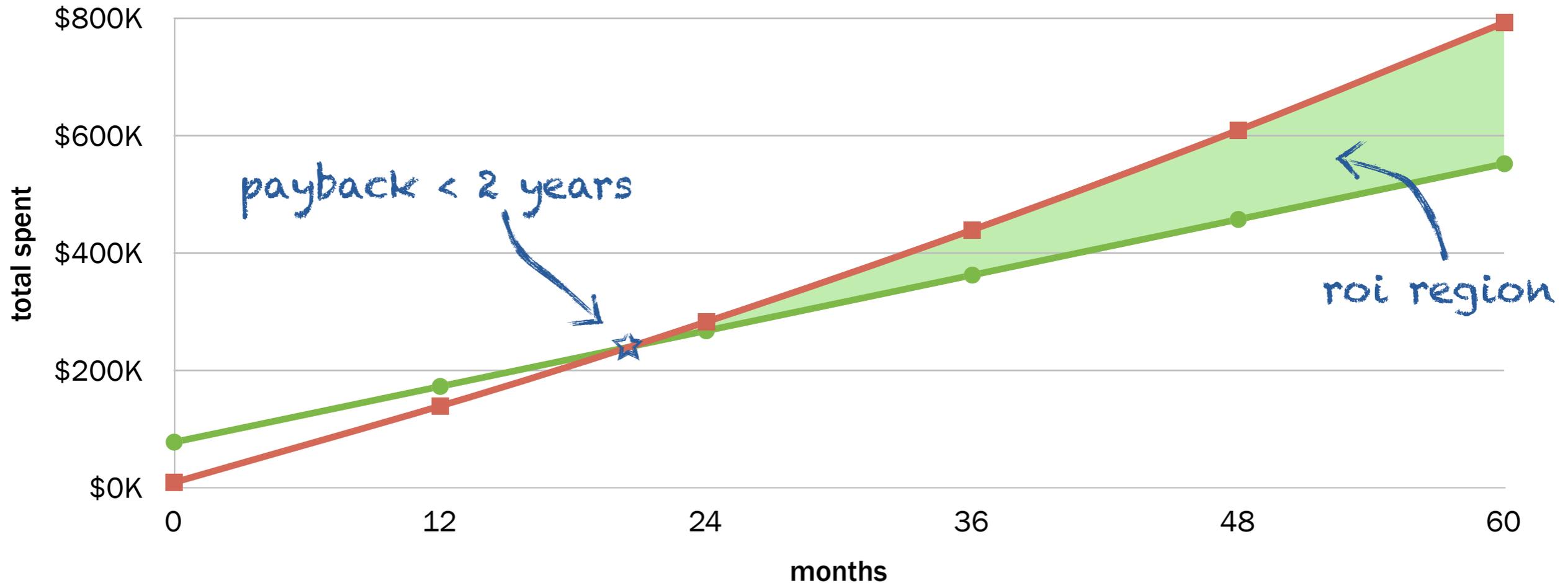
increased customer satisfaction

communications cost savings



# roi / tco analysis

■ incumbent hosted voip      ● new uc solution



## each price includes

- ➡ MRC network and voice PSTN charges
- ➡ NRC costs for addition of 24 users/yr.

## each price does NOT include

- ➡ “soft-cost” productivity gains
- ➡ travel savings impact

- ▶ **crm integration**
  - click-to-call and screen pop for sales team
- ▶ **mobile voip (FMC)**
  - SIP, VoWiFi, eventually Vo3G/Vo4G
  - use of smartphones as primary office extension
  - field testing a couple of different solutions
- ▶ **communications enabled business processes**
  - context-aware IM, voice, contact list (presence)
  - field workers have quick access to knowledge workers via multiple communication methods



# **5 tips** for building a distributed uc/sip architecture

# 1

- ▶ make sure your LAN and WAN is voip/uc ready
  - understand the (significant) infrastructure requirements that are necessary to support uc
  - LAN: architecture, duplex, capacity
  - WAN: QoS, adequate bandwidth for growth, low latency, jitter, packet loss
    - Make sure it actually works (tested!)
  - network is number one culprit of poor uc performance and adoption



- ▶ **PathViewCloud**  by **Apparent**  NETWORKS™
  - SaaS network monitoring / assessment
  - uses one or more “sequencers” on LAN
- ▶ continuous testing of VoIP-sensitive network metrics
  - jitter
  - packet-loss
  - latency
  - QoS tagging and integrity
- ▶ test MPLS and internet destinations
  - verify network readiness *before* deployment
  - monitor capacity *after* installation (sp always changing)

# network assessment

Group: VoIP (15 paths) 15

↑ Name	Network Type	Target Type	Importance	Last Diagnostic	Violations Today / Past 7 Days	Action
Colo SG220 (Remote Polycoms) (192.168.200.4)	LAN	Voice Server: Shoretel	8	01/13/2011 17:33	0 / 0	<input type="button" value="Action"/>
Colo SG90 (SIP Trunks) (192.168.200.5)	LAN	Voice Server	8	01/13/2011 17:34	0 / 0	<input type="button" value="Action"/>
Colo VoIP Internet Edge (██████████)	WAN	Voice WAN	8		0 / 39	<input type="button" value="Action"/>
Corona SG50 (192.168.8.20)	WAN	Voice WAN	8	01/27/2011 07:02	0 / 1	<input type="button" value="Action"/>
El Seg to Colo MPLS (no QoS) (192.168.200.80)	WAN	Voice Server: Shoretel	8		0 / 2	<input type="button" value="Action"/>
El Segundo SG90 (192.168.5.2)	WAN	Voice WAN	8	01/17/2011 19:20	0 / 0	<input type="button" value="Action"/>
Frank Romeo (NJ ext 220) (██████████)	WAN	Voice WAN	5	01/30/2011 20:33	0 / 2	<input type="button" value="Action"/>
Laguna SG50 (192.168.11.20)	WAN	Voice WAN	8	01/17/2011 19:19	0 / 0	<input type="button" value="Action"/>
MELLISA DAHL (██████████)	WAN	Voice WAN	5	01/30/2011 20:21	0 / 3	<input type="button" value="Action"/>
New Jersey SG50 (192.168.9.20)	WAN	Voice WAN	8	01/17/2011 18:49	0 / 0	<input type="button" value="Action"/>
New York Office (ext 303) (██████████)	WAN	Voice WAN	5	02/02/2011 10:08	4 / 17	<input type="button" value="Action"/>
North Carolina SG50 (192.168.12.20)	WAN	Voice WAN	8	01/22/2011 06:00	0 / 0	<input type="button" value="Action"/>
Plano SG50 (192.168.10.20)	WAN	Voice WAN	8	01/31/2011 09:29	1 / 9	<input type="button" value="Action"/>
SIParator (internal) (192.168.200.10)	LAN	Voice Server	8	01/13/2011 17:09	0 / 0	<input type="button" value="Action"/>
SIParator (Internet Edge) (██████████)	WAN	Voice WAN	7	02/01/2011 13:50	0 / 37	<input type="button" value="Action"/>

# network assessment

## MOS

Most Recent: 4.4

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
MOS	4.4	4.4	4.4

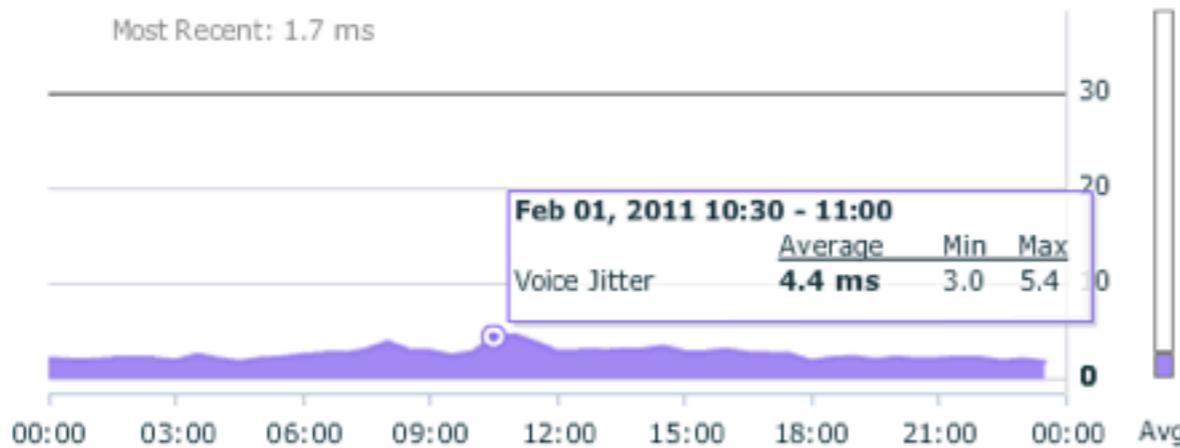


## Voice Jitter

Most Recent: 1.7 ms

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Voice Jitter	4.4 ms	3.0	5.4

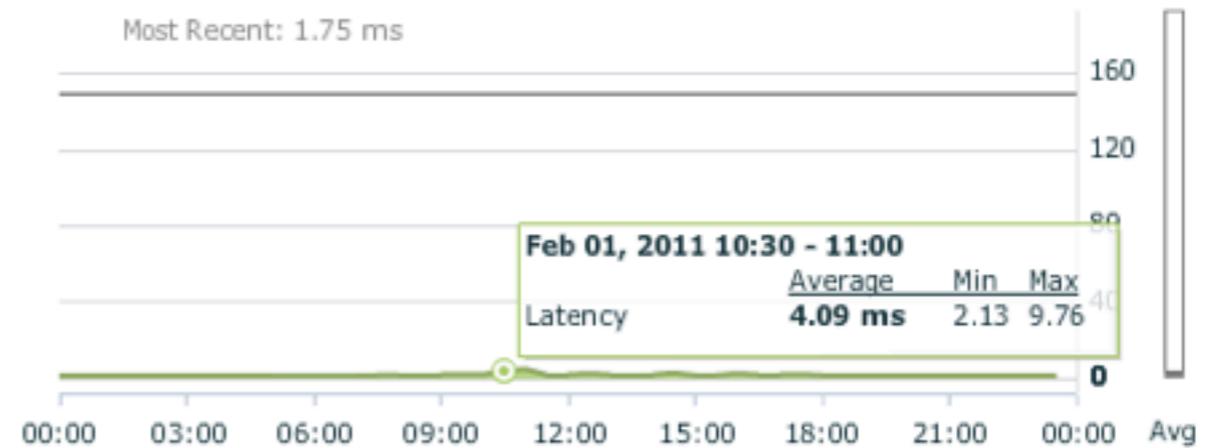


## Latency

Most Recent: 1.75 ms

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Latency	4.09 ms	2.13	9.76

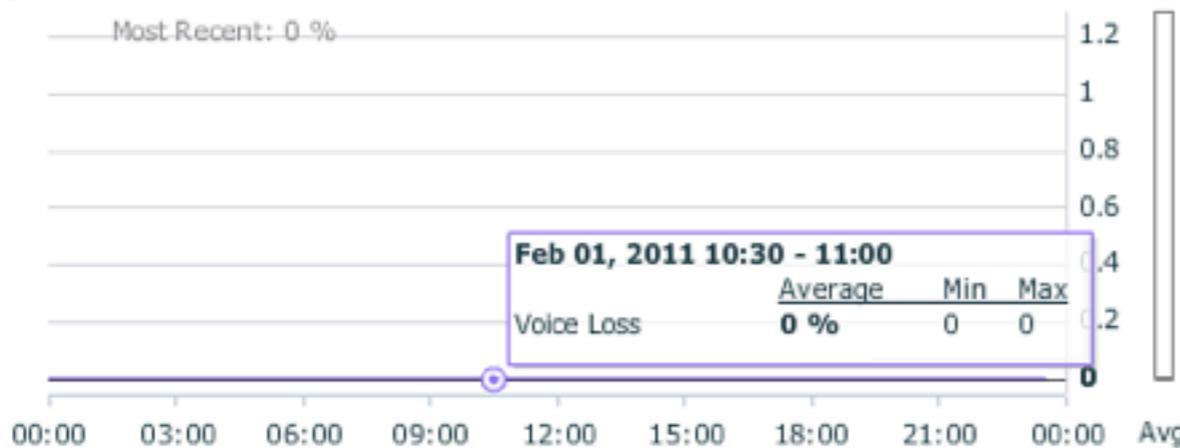


## Voice Loss

Most Recent: 0 %

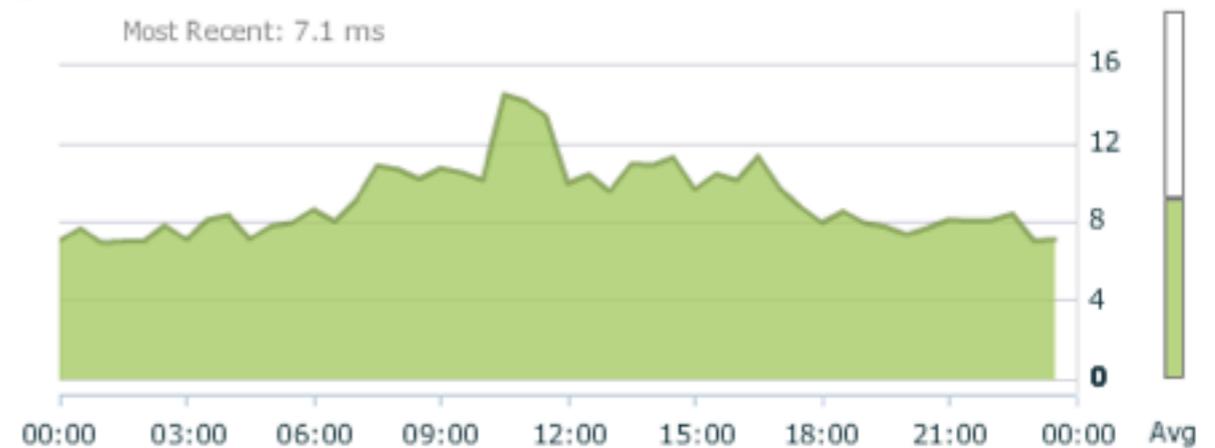
Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Voice Loss	0 %	0	0

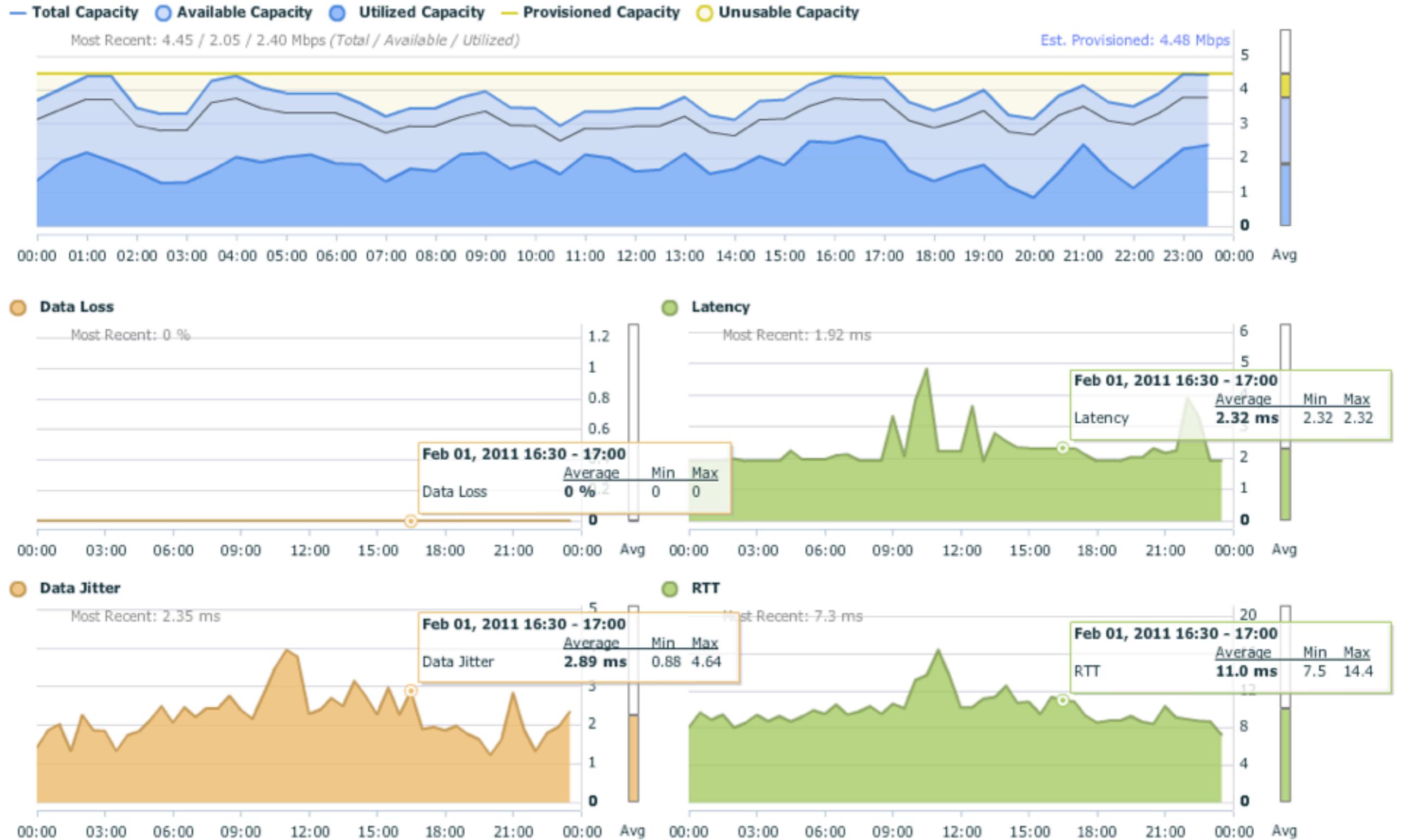


## RTT

Most Recent: 7.1 ms



# network assessment



# network assessment

unwired

Summary Voice Detail

Hide Diagnostics

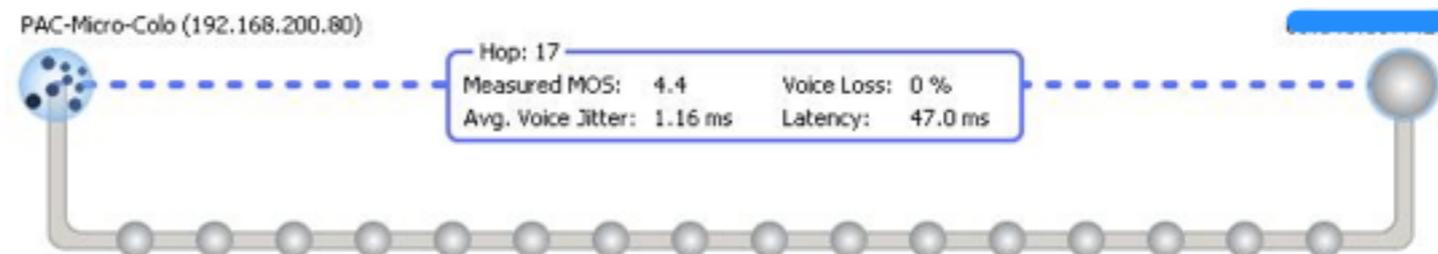
Hop	Severity	IP Address	Host Name	Voice Loss (%)	Measured MOS	Best MOS	Worst MOS	Latency (ms)	Voice Jitter (Avg/Max, ms)	RTT (Min/Avg/Max, ms)	QoS (Set/Measur)	
1	✓	192.168.200.1	192.168.200.1	0	4.4	4.4	4.4	0.31	0.20 / 4.51	0.63 / 0.80 / 1.40		
2	!	[REDACTED]	[REDACTED].ip.mcleodusa.net	0	4.3	4.3	3.8	0.25	3.63 / 106	0.50 / 7.23 / 118		
		Frequency	Observation									
		!	Inconsistent handling of voice packets detected [ Under network load, voice applications may experience excessive jitter ] [ Check for tendency toward decreasing MOS with increasing call load ]									
3	✓	66.251.35.210	gi-4-0-0-13.core02.anhmca01.paetec.net	0	4.4	4.4	4.4	0.19	0.08 / 0.87	0.39 / 0.50 / 1.02		
4	✓	64.80.253.213	po-3-0-0.core01.lsa ca01.paetec.net	0	4.4	4.4	4.4	1.81	0.07 / 0.26	3.61 / 3.73 / 3.97		
5	✓	66.251.30.5	gi-2-0-4.gw02.lsa ca01.paetec.net	0	4.4	4.4	4.4	1.84	0.09 / 0.48	3.67 / 3.79 / 4.11		
6	✓	207.138.128.165	ge-5-2-0.406.ar3.lax1.gblx.net	0	4.4	4.4	4.4	1.93	2.17 / 107	3.87 / 3.97 / 4.24		
7	✓	64.208.17.198	comcast-ip-services-llc.tengigabithernet3-4.ar4.lax2.gblx.net	0	4.4	4.4	4.4	39.2	0.43 / 8.76	78.4 / 79.0 / 80.0		
8	i	68.86.85.142	pos-0-14-0-0-cr01.dallas.tx.ibone.comcast.net	0	4.4	4.4	4.4	20.2	0.23 / 10.9	40.5 / 40.8 / 60.4		
		Frequency	Observation									
		i 21%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
9	i	68.86.85.222	pos-0-11-0-0-cr01.atlanta.ga.ibone.comcast.net	0	4.4	4.4	4.4	27.8	0.31 / 17.1	55.7 / 56.0 / 56.9		
		Frequency	Observation									
		i 11%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
10	✓	68.86.87.193	pos-1-9-0-0-cr01.ashburn.va.ibone.comcast.net	0	4.4	4.4	4.4	38.6	0.14 / 5.13	77.2 / 77.4 / 77.9		
11	i	68.86.95.158	68.86.95.158	0	4.4	4.4	4.4	40.9	1.66 / 28.1	81.8 / 82.1 / 111		
		Frequency	Observation									
		i 9%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
12	i	68.85.62.189	te-0-6-0-6-ar01.audubon.nj.panjde.comcast.net	0	4.4	4.4	4.4	41.0	0.23 / 1.22	82.0 / 82.5 / 83.5		
		Frequency	Observation									
		i 13%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
13	✓	68.85.35.10	te-9-8-ar01.absecon.nj.panjde.comcast.net	0	4.4	4.4	4.4	41.8	0.77 / 28.6	83.6 / 84.5 / 99.7		

# network assessment

14		68.85.159.14	te-2-1-ar01.eatontown.nj.panjde.comcast.net	0	4.4	4.4	4.4	41.3	1.77 / 22.6	82.5 / 85.6 / 94.9	- / -	
		Frequency	Observation									
		24%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
15		68.86.210.154	te-1-1-ur01.middlestown.nj.panjde.comcast.net	0	4.4	4.4	4.4	41.5	1.54 / 11.4	83.0 / 86.0 / 89.4	- / -	
		Frequency	Observation									
		32%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
16		68.87.214.214	68.87.214.214	0	4.4	4.4	4.4	41.5	1.51 / 3.17	82.9 / 86.2 / 88.8	- / -	
		Frequency	Observation									
		28%	Packet reordering detected [ Relatively low levels should not affect network performance ]									
17		69.141.187.42		0	4.4	4.4	4.4	47.0	1.16 / 12.3	94.1 / 97.0 / 130	- / -	
		Frequency	Observation									
		33%	Packet reordering detected [ Relatively low levels should not affect network performance ]									

## MELLISA DAHL

**Target Type:** Voice WAN  
**Sequencer:** PAC-Micro-Colo  
**Target:**   
**Last Diagnostic:** 02/02/2011 10:46



## 2

- ▶ use sip trunking unless you have a good reason not to
  - pre-qualify pbx and service provider for interoperability
  - shoot for native SIP, but use SIP to TDM gateway if you must to get by for now
  - foundation for next decade of communications
  - be careful about OTT ITSPs due to unpredictable links
  - plan for a learning curve...



## 3

- ▶ understand how your business departments and users communicate and collaborate
  - be methodical and as objective as possible
  - use surveys, evaluate IT/IS systems, user categorization
  - take quantitative measurements of task efficiencies (create a baseline)
  - determine “hot spots” where improved communication can make a measurable difference



## ▶ evaluate multiple uc vendors

- attend seminars, watch demos, read white papers
- rate strong and weak points of each
- determine how well each delivers on your “hot spots”
- generally the larger the organization, the more vendors required to build solution

# 4



# 5

- ▶ phase-in the deployment of your uc systems
  - consider implementation specialists
  - highest value at least cost pieces first
  - pilot and test each phase as much as possible
  - provide end-user training
  - test test test before rollout
  - don't forget security



- ▶ sip trunking is the glue of the evolved pbx/ucc
  - inGate SIParator is the tube



**today**

ties together uc components

**tomorrow**

ties together uc systems



- ▶ business is evolving, so must your communications infrastructure
  - TDM won't cut it in the long term...
- ▶ the evolved pbx uses SIP for trunks AND endpoints
- ▶ the network is more fragile than uc systems / components themselves
- ▶ “full uc” is non-trivial, usually involves multiple vendors and phases
- ▶ an integration partner is very valuable, in many cases, a must

# thank you!



- ▶ question & answer
- ▶ <http://www.unwiredrevolution.com>
- ▶ [mvlasach@unwiredrevolution.com](mailto:mvlasach@unwiredrevolution.com)