

SIP: Connecting the Distributed Workforce

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ITEXPO East
February 3, 2011

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

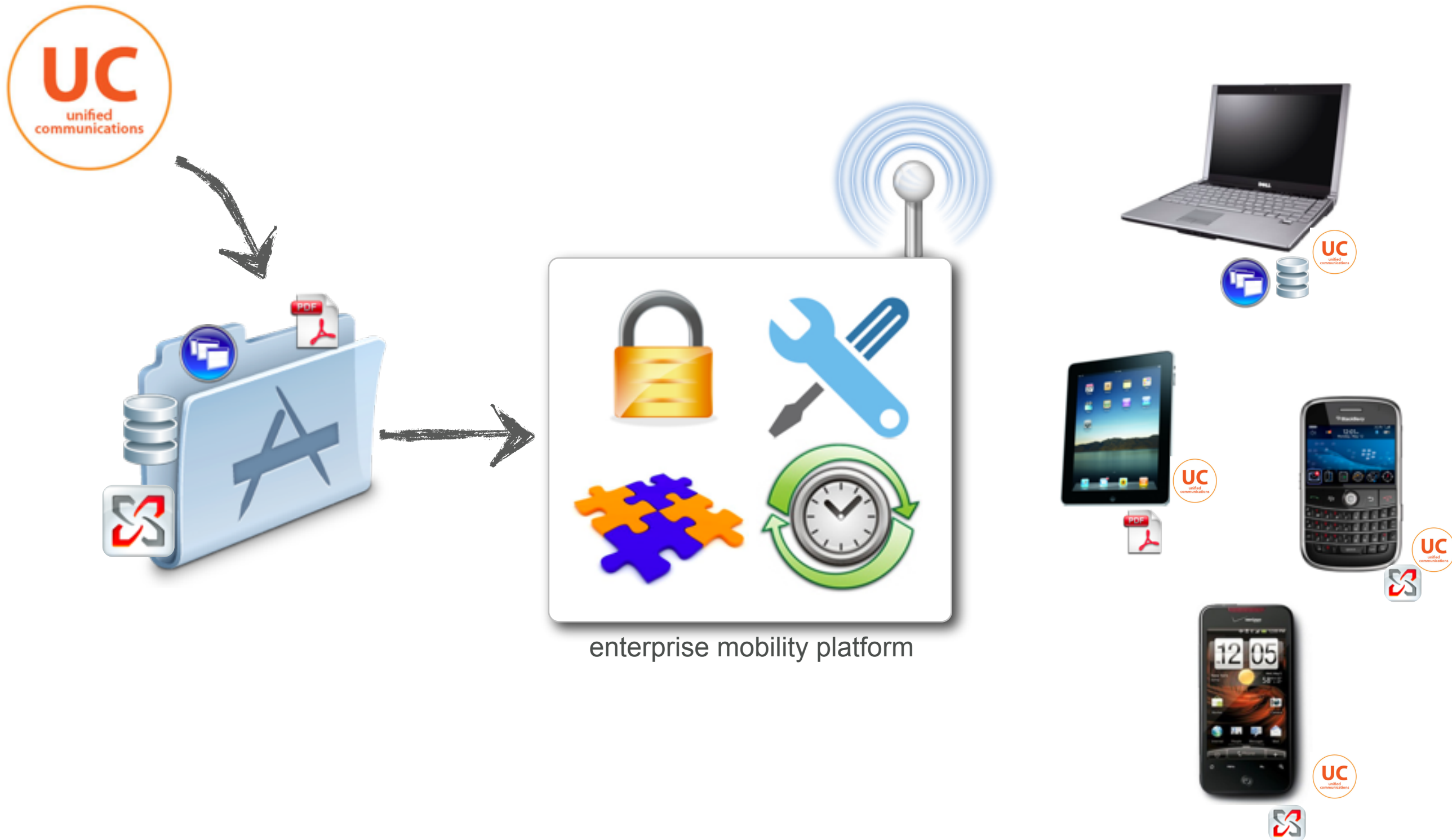
customers

unwired
revolution



mobilizing uc

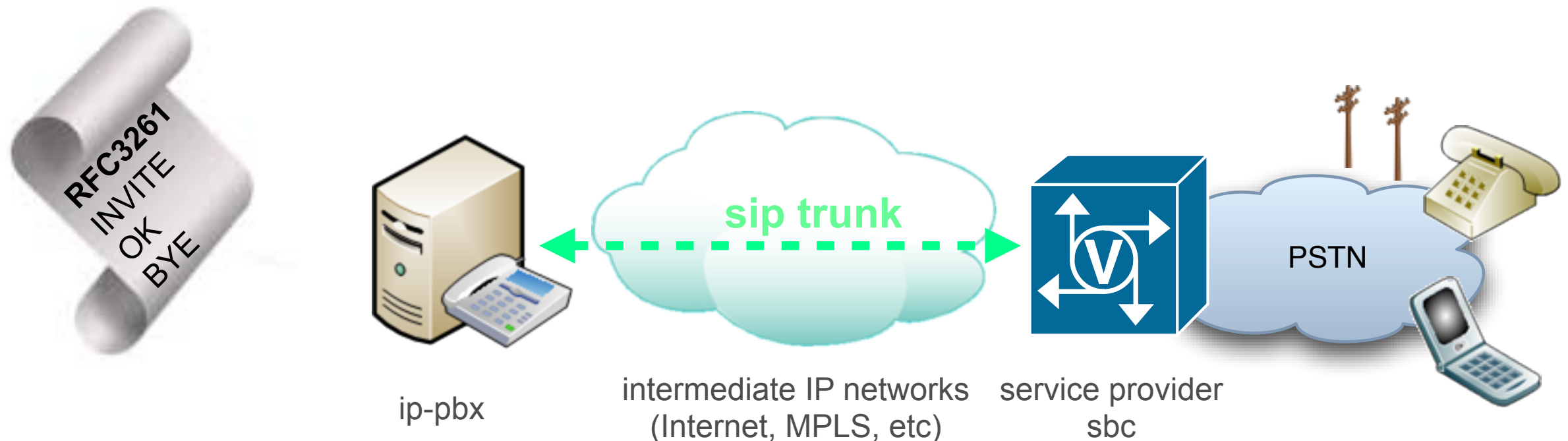
unwired
revolution



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













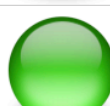
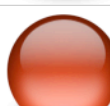
presence

- ▶ 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>

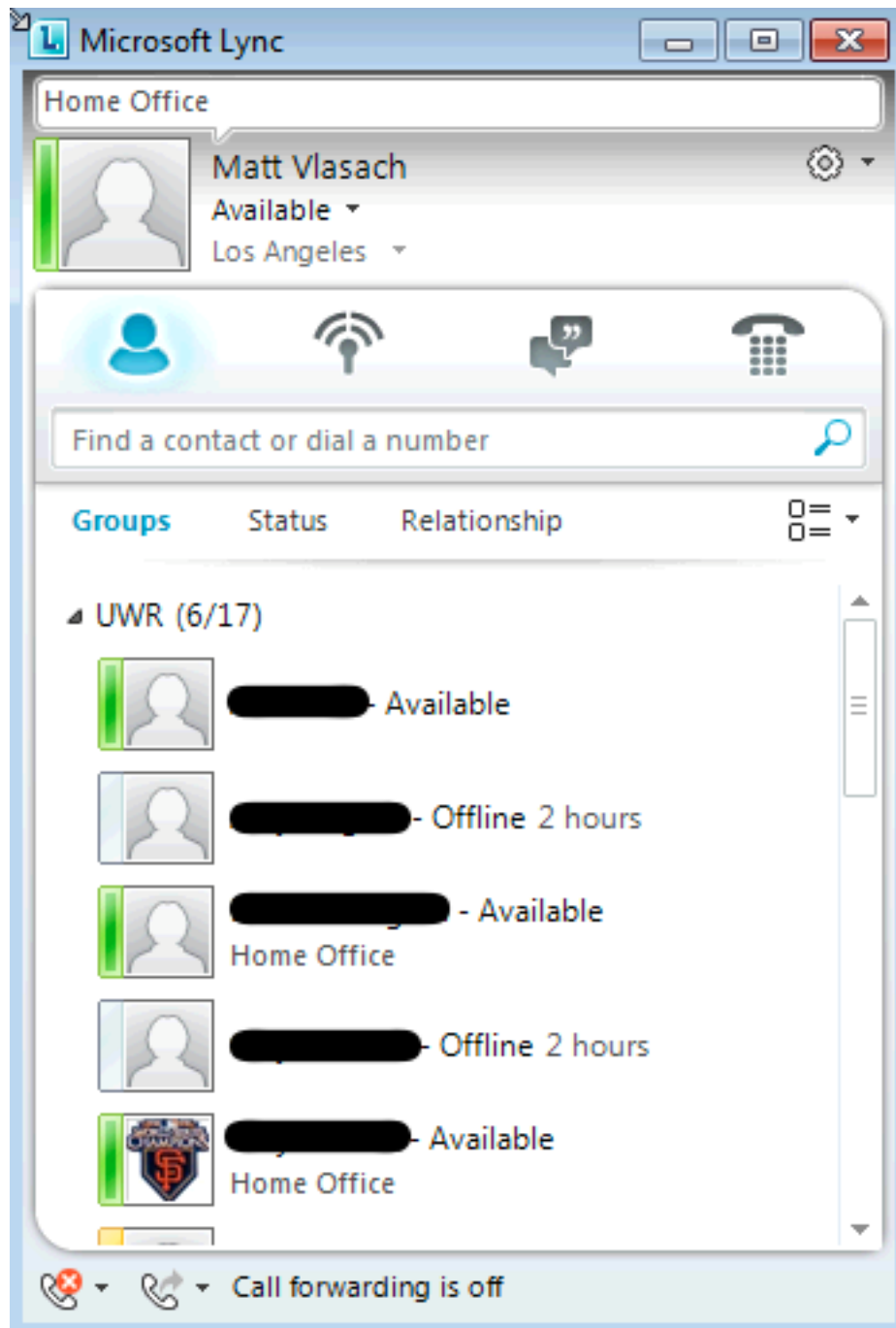
- ▶ all vendors have different strengths/weaknesses
 - different backgrounds, priorities, dev cycles
 - no single vendor makes the silver bullet uc solution
 - perpetuated by continuous evolution of ucc tech

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

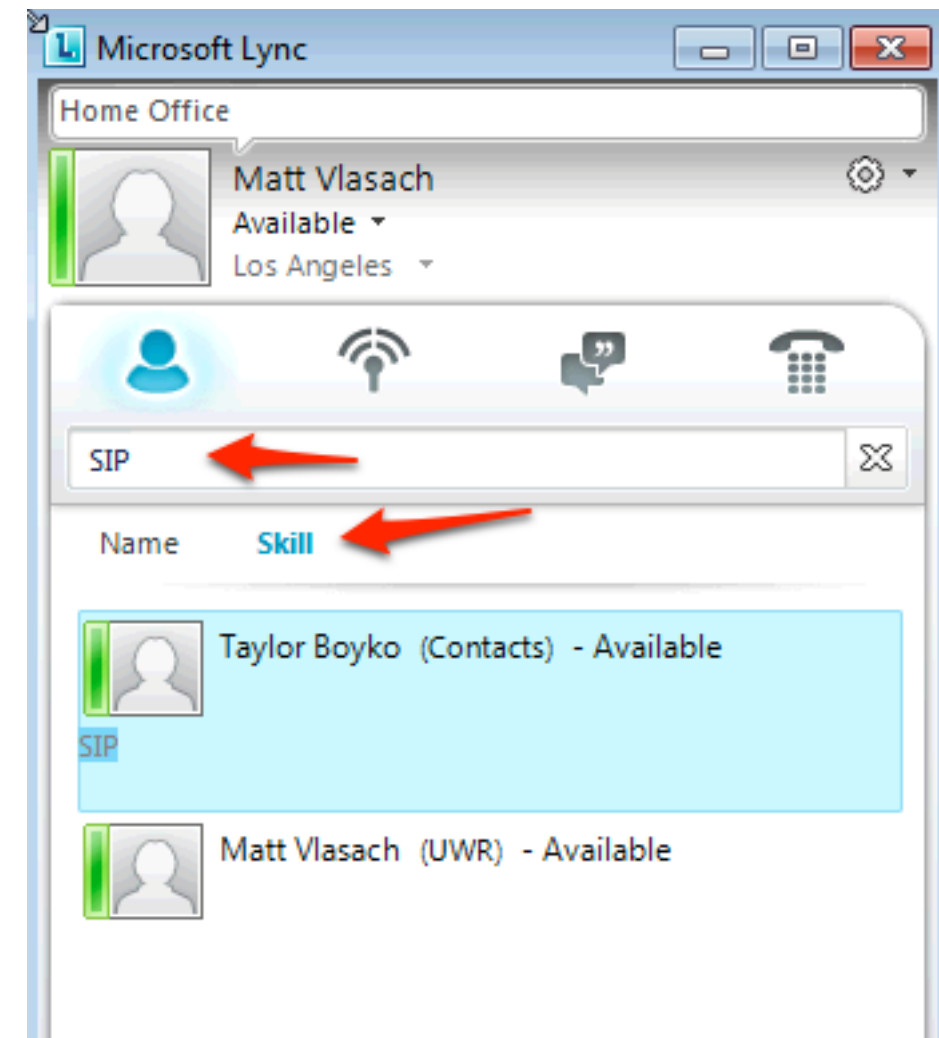
example: lync 2010

- ▶ 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

The screenshot shows the 'Untitled - Meeting' window in Microsoft Outlook. The ribbon includes 'File', 'Meeting', 'Insert', 'Format Text', and 'Review'. The 'Meeting' tab is active, showing options like 'Delete', 'Calendar', 'Forward', 'OneNote', 'Appointment', 'Scheduling Assistant', 'Join Online Meeting', 'Meeting Options', 'Cancel Invitation', 'Attendees', 'Options' (with 'Busy' and '10 minutes' selected), 'Recurrence', 'Time Zones', 'Room Finder', 'Categorize', 'Tags', 'Zoom', and 'Zoom'.

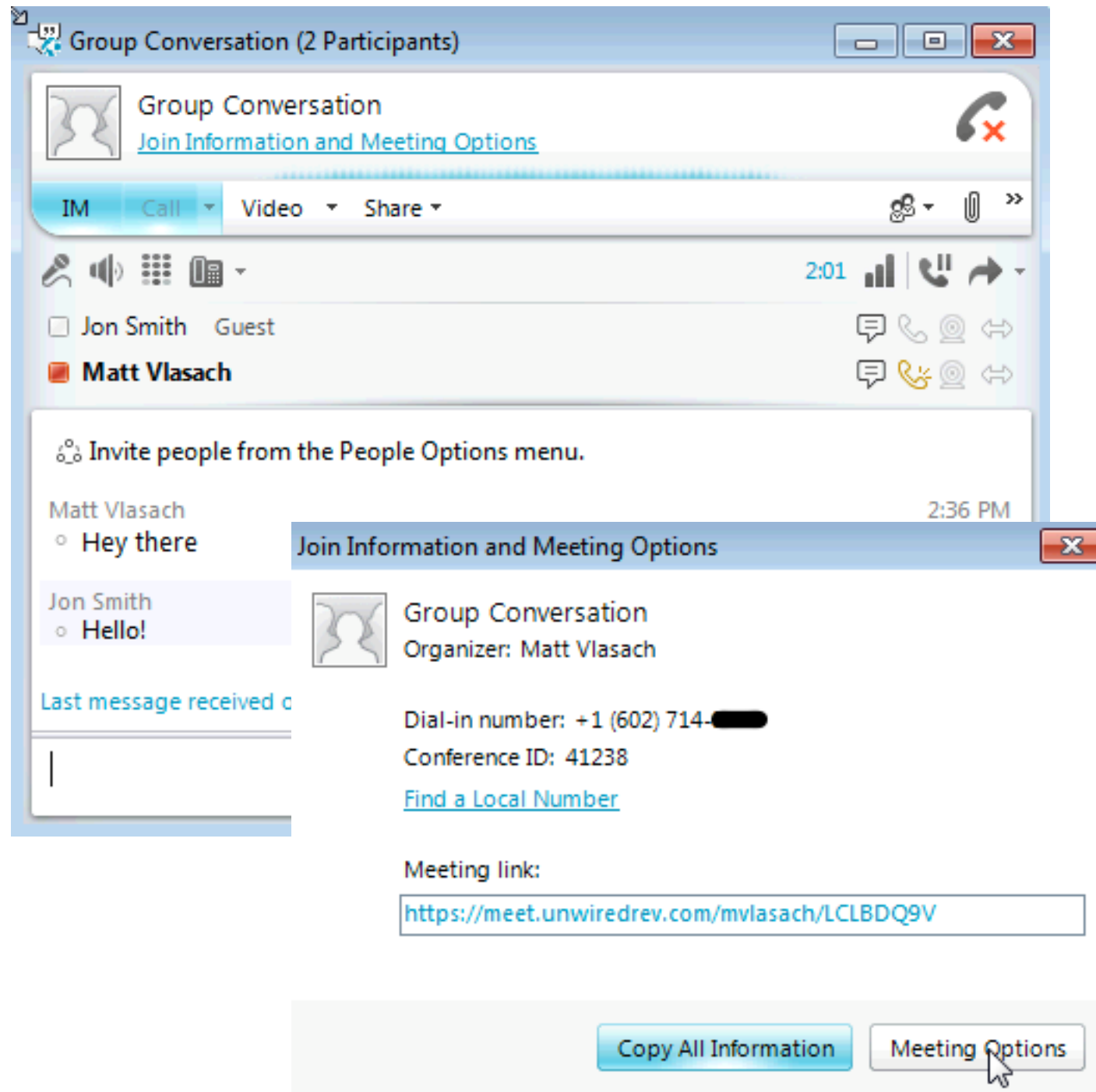
Below the ribbon, a message states: 'Invitations have not been sent for this meeting.' The 'From' field is 'MV Unwired'. The 'To' field is highlighted with a red box and contains four redacted email addresses. The 'Subject' is 'Discuss Project'. The 'Location' is 'Online Meeting'. The 'Start time' is 'Mon 2/14/2011 12:00 AM' and the 'End time' is 'Mon 2/14/2011 12:30 AM'. There is an 'All day event' checkbox.

The meeting details section includes a link to 'Join online meeting' with the URL <https://meet.unwiredrev.com/mvlasach/BQD3SC80>. Below this is 'Join by Phone' with the number '+1 (602) 714-XXXX' and a link to 'Find a local number'. The 'Conference ID' is '58905'. Two red arrows point to the phone number and the 'Find a local number' link.

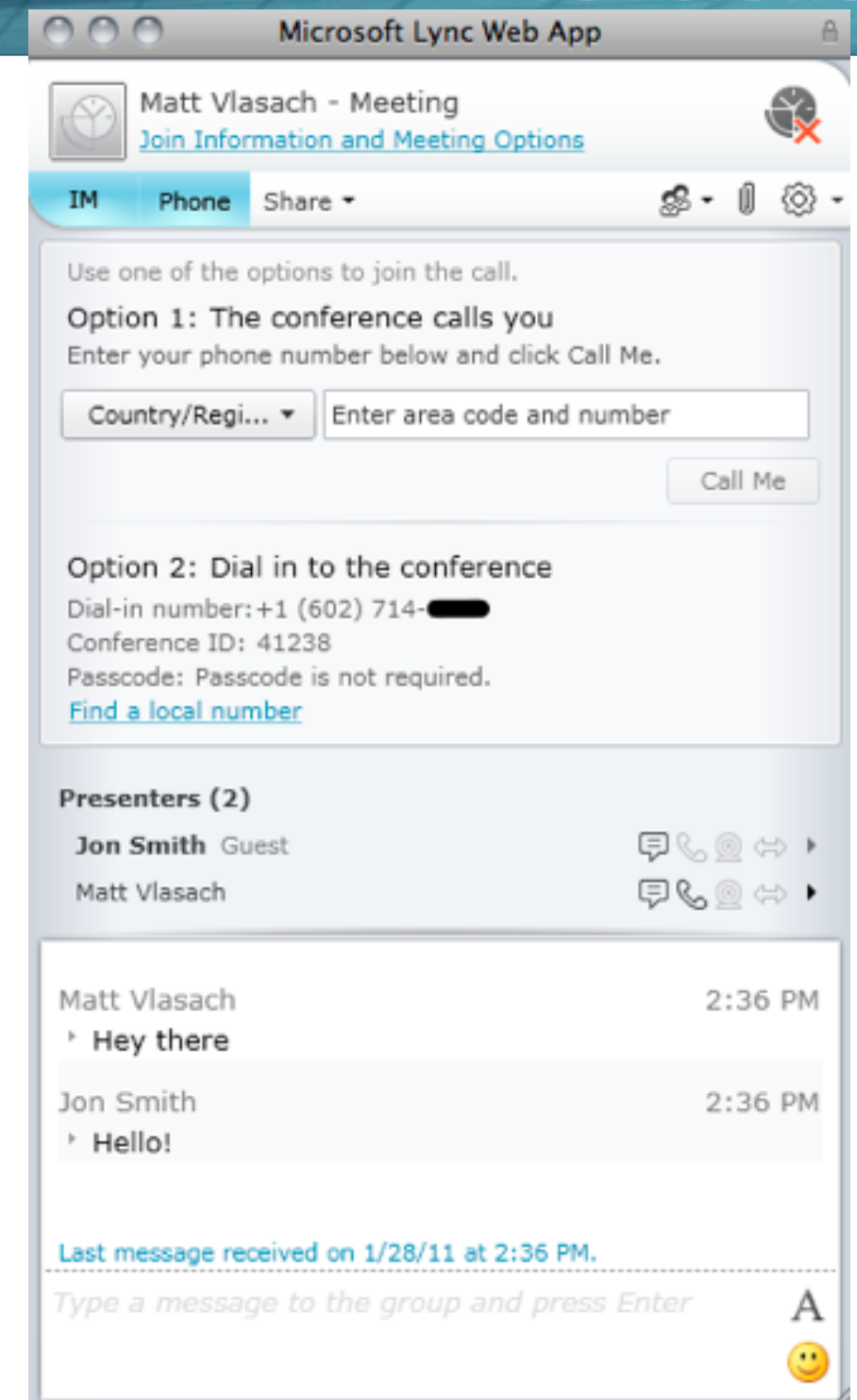
At the bottom, it says 'In Shared Folder: Calendar' and shows a list of participants, including 'Matt Vlasach'.

outlook integration, meeting scheduling

lync 2010

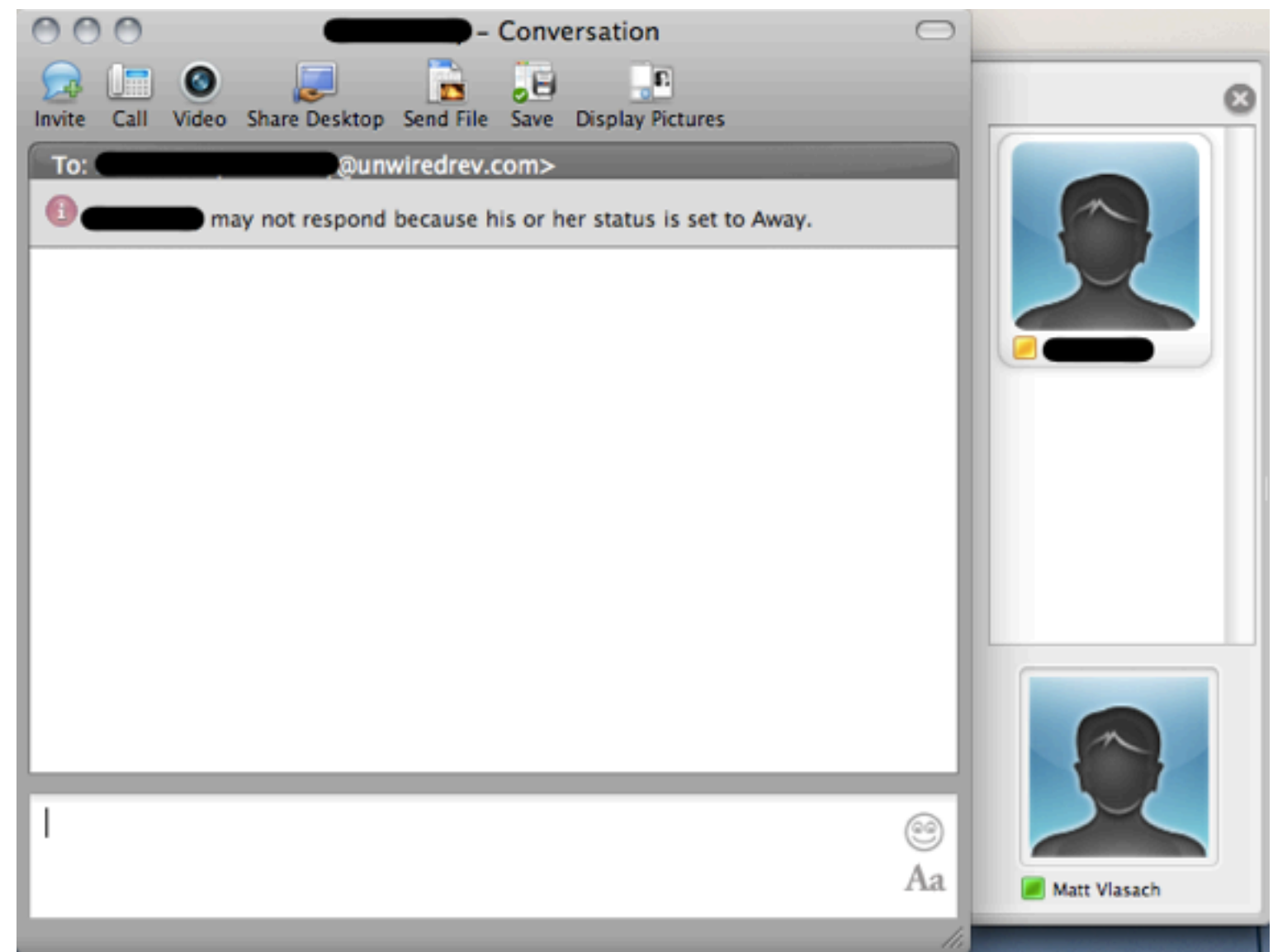
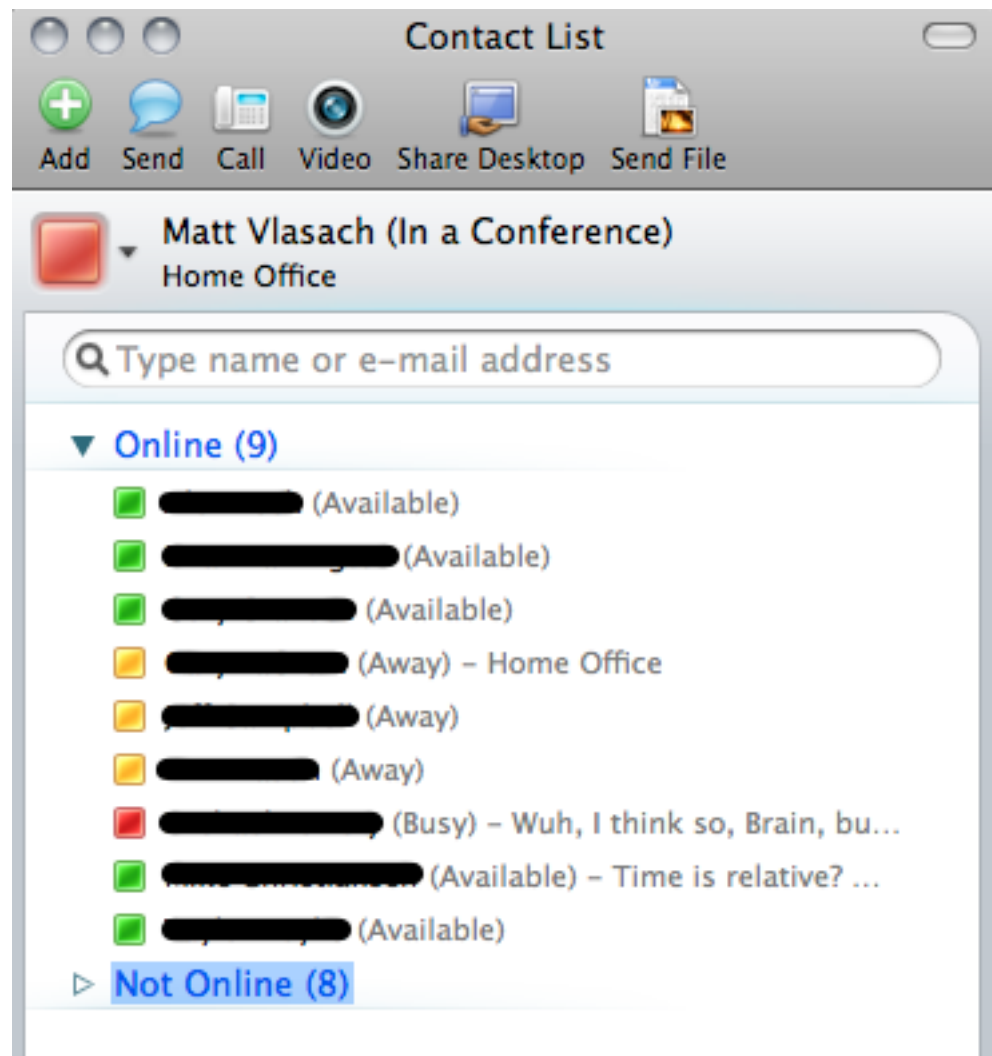


lync “conversation” window



“lync web app” silverlight viewer

lync 2010

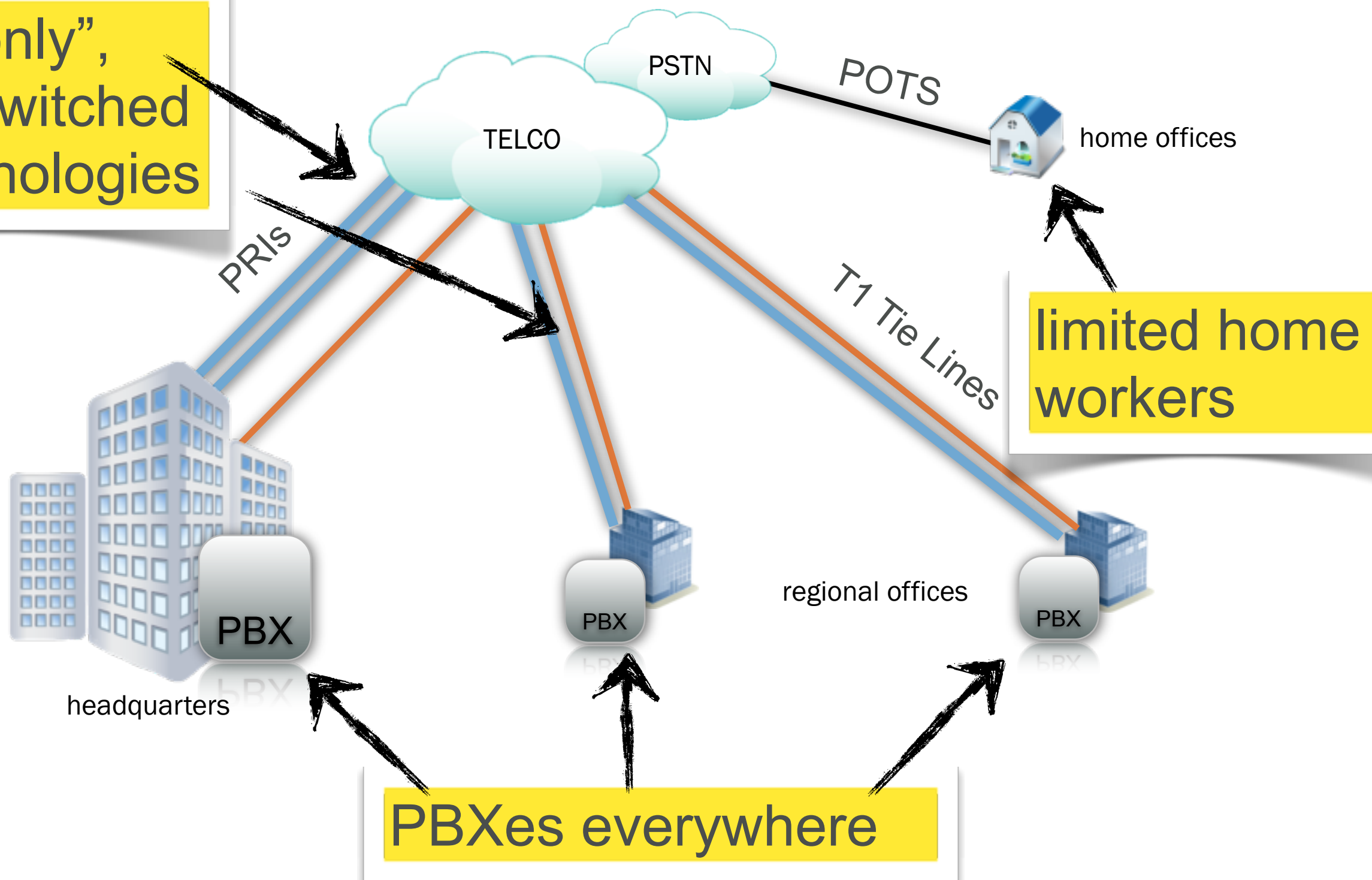


mac communicator

the **evolved** pbx architecture

legacy architecture

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



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

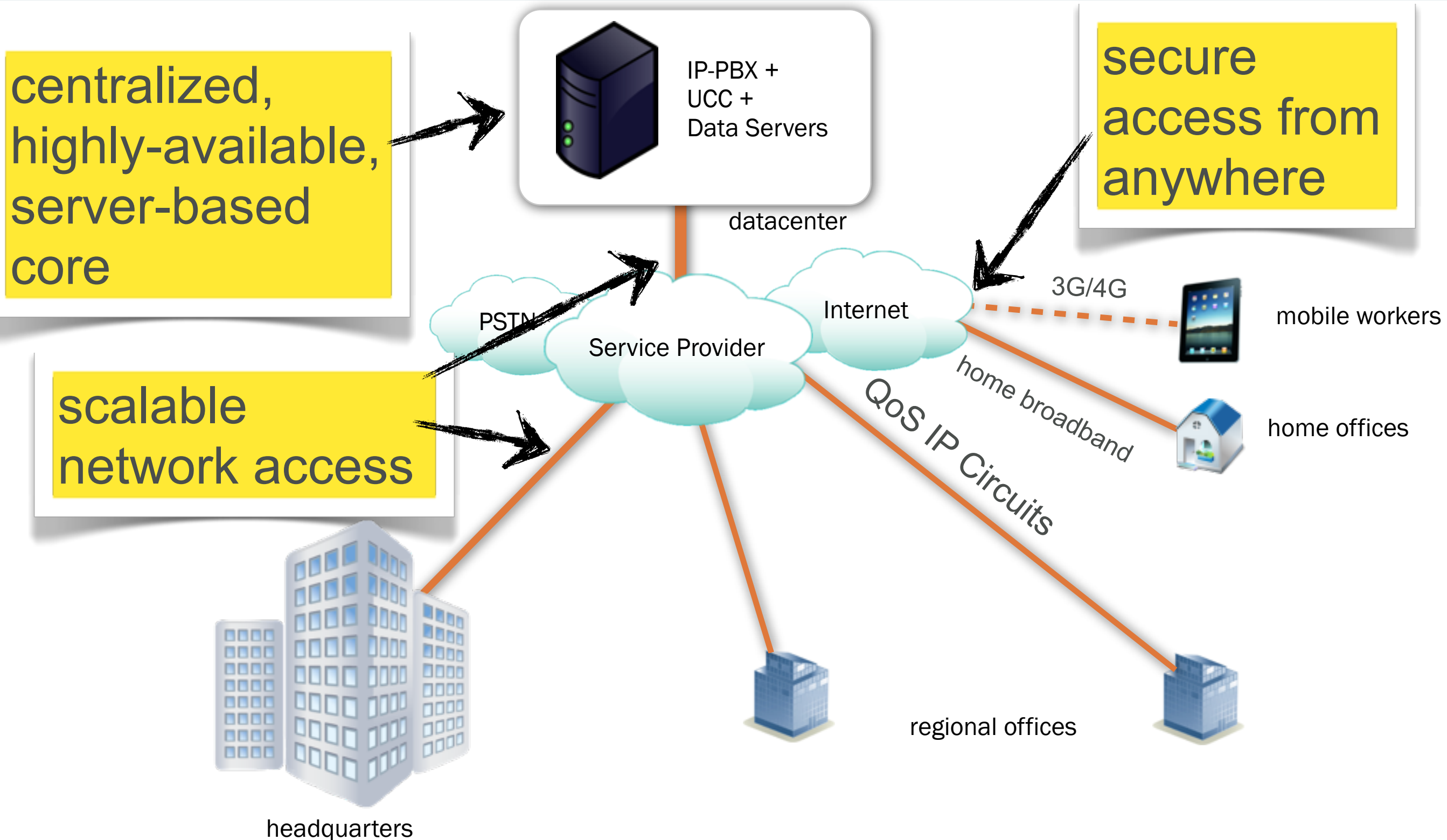
highly
distributed &
collaborative
workforce

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



all connected by sip!



case study

partner engineering and science, inc.

about partner esi



- ▶ 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

centralized trunking

it infrastructure requirements

user applications and services

high availability

scaleable

fixed-remote voip

solution components



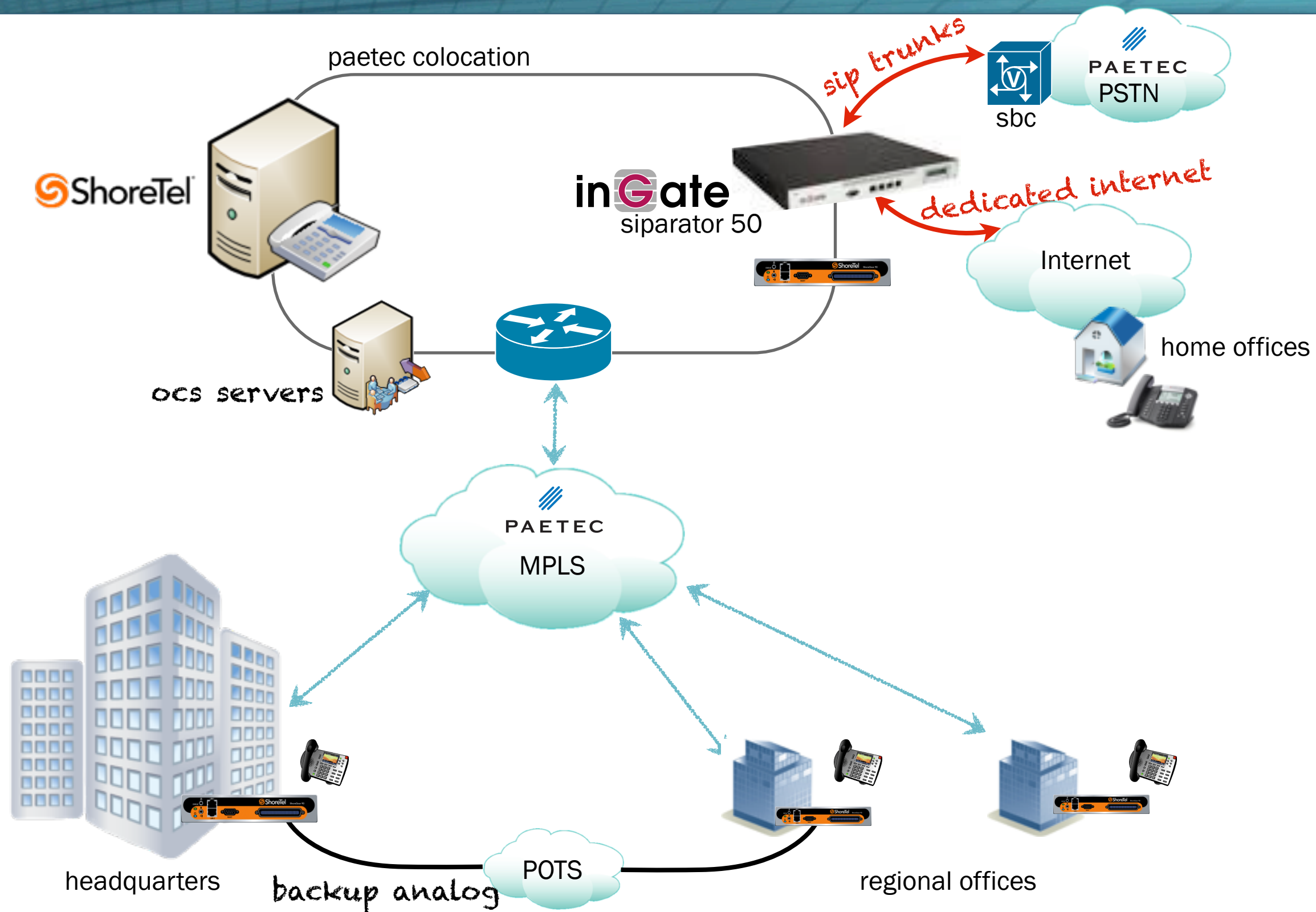
Lync 2010



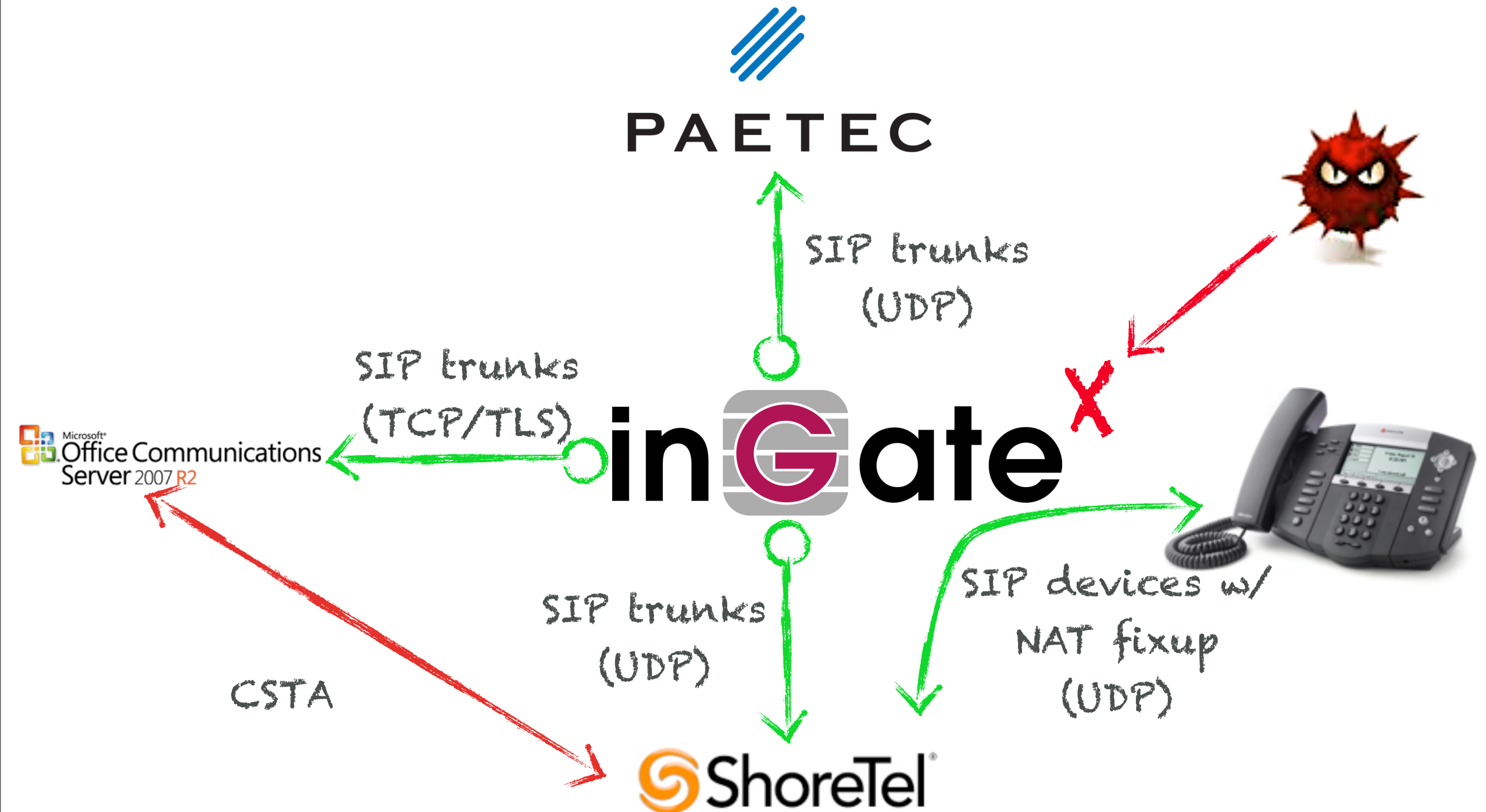
on-net SIP trunking	voip telephony + unified messaging	sip trunking interoperability	im + presence	SIP-compliant endpoints
mpls network with qos	find me / mobility support	remote SIP device (NAT fixup)	office integration with click-to-call	centralized provisioning support
colocation with ethernet handoffs	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

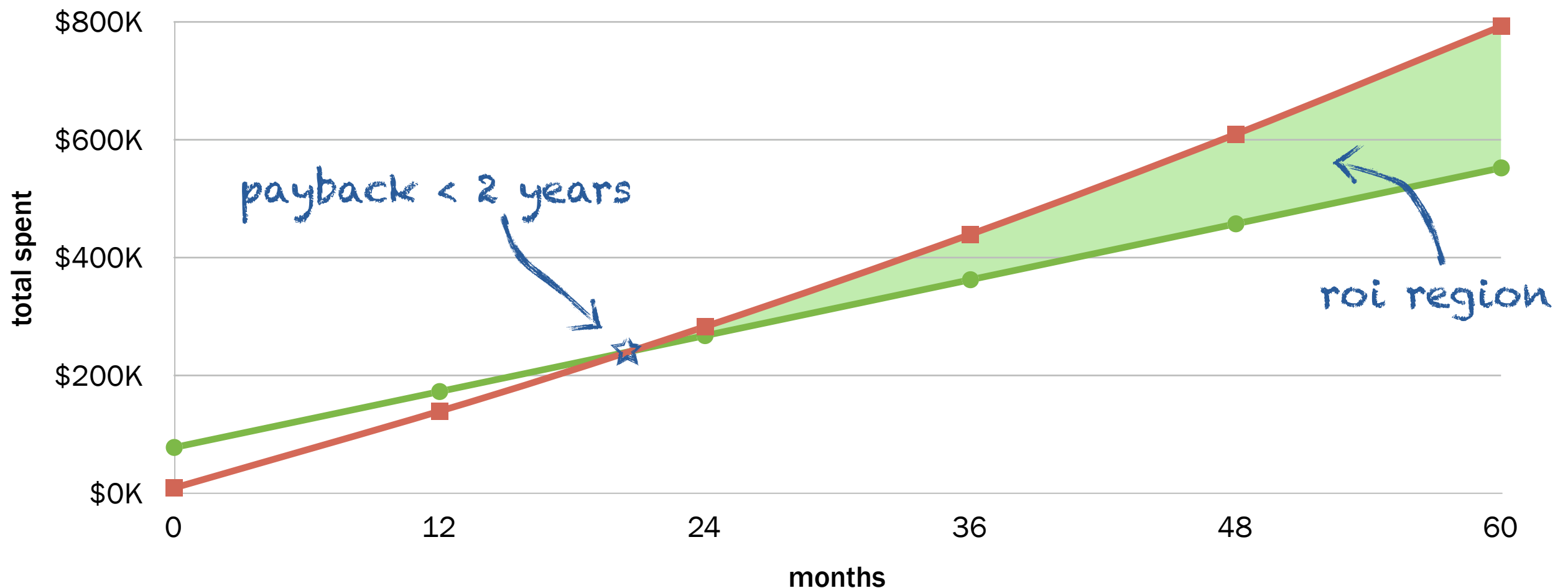
increased customer satisfaction

communications cost savings



roi / tco analysis

■ incumbent hosted voip ● new uc solution



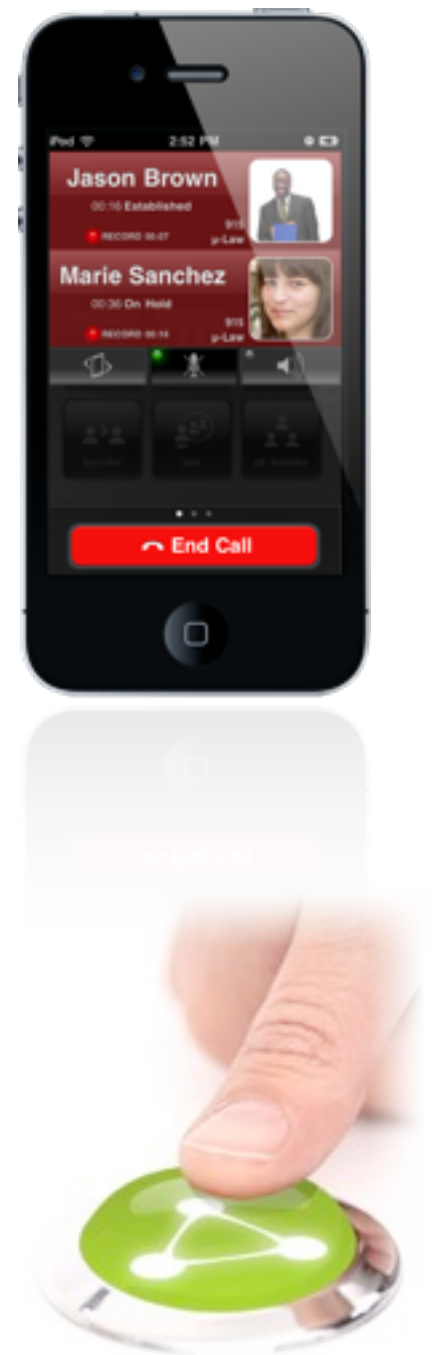
each price includes

- ➡ 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




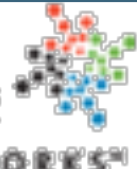
5 tips for building a distributed uc/sip architecture

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



network assessment

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

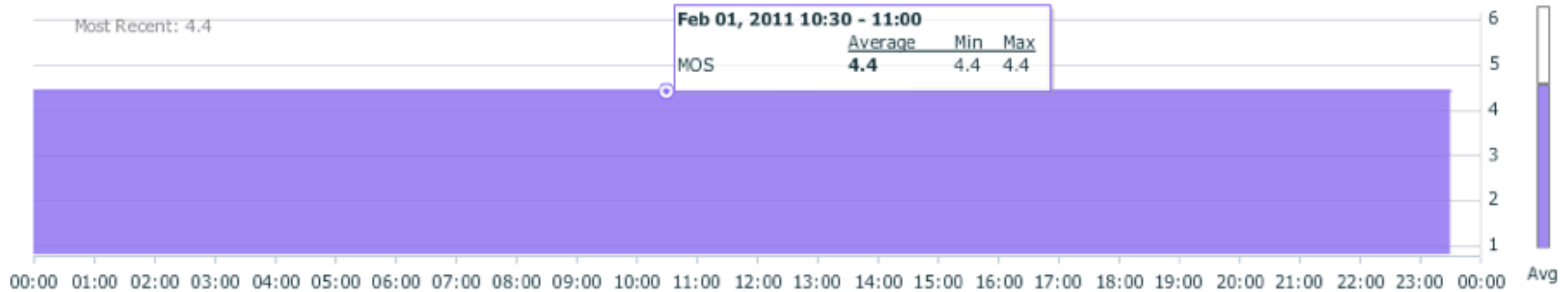
network assessment

MOS

Most Recent: 4.4

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
MOS	4.4	4.4	4.4

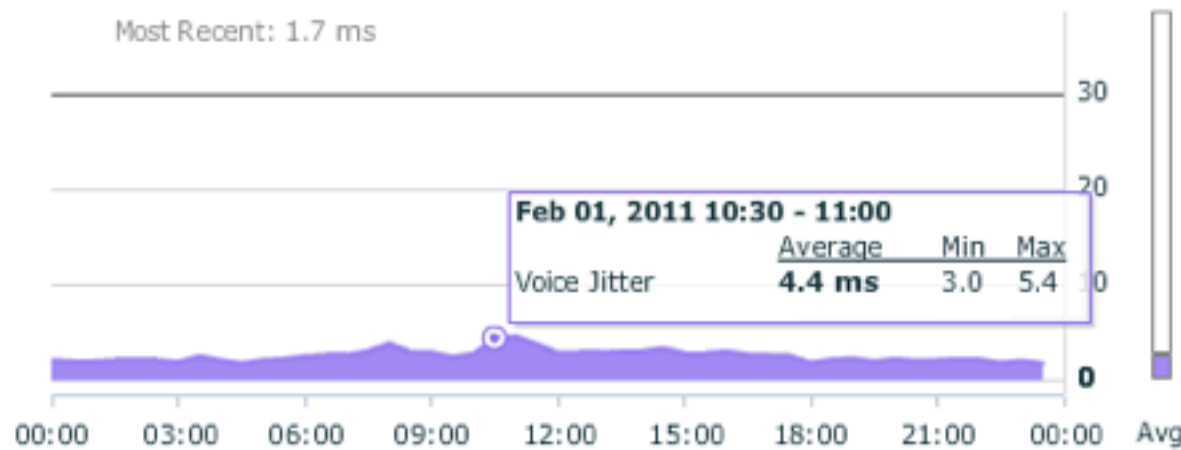


Voice Jitter

Most Recent: 1.7 ms

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Voice Jitter	4.4 ms	3.0	5.4

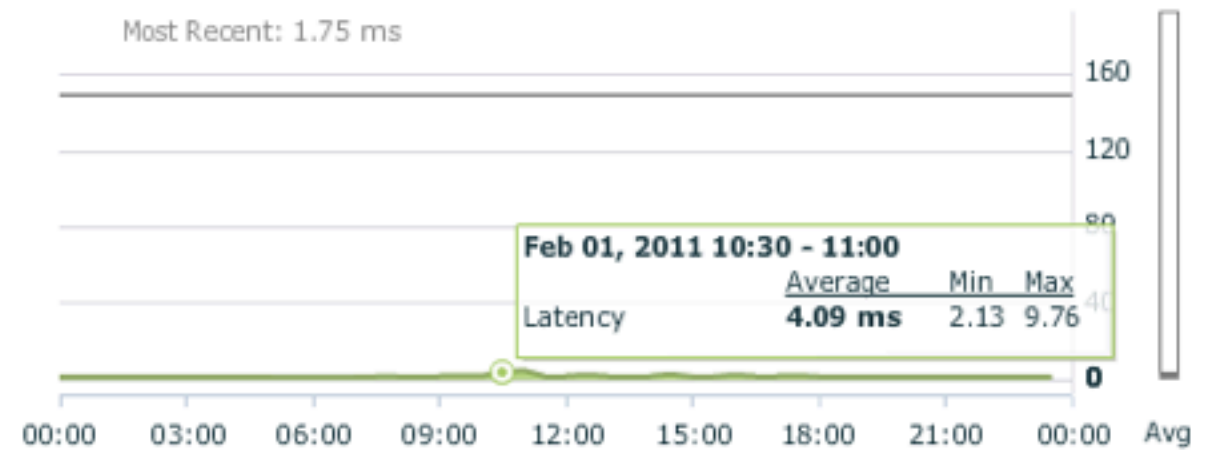


Latency

Most Recent: 1.75 ms

Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Latency	4.09 ms	2.13	9.76

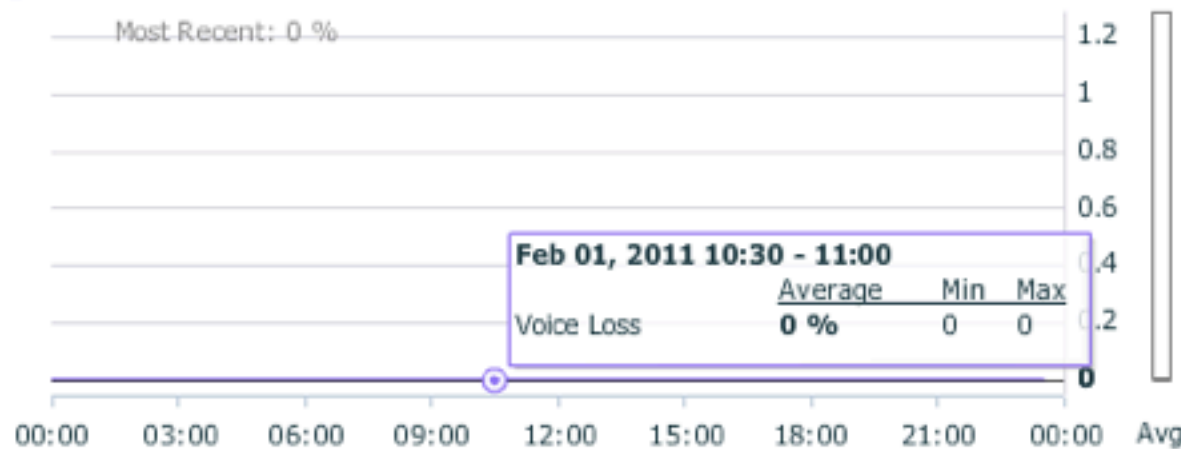


Voice Loss

Most Recent: 0 %

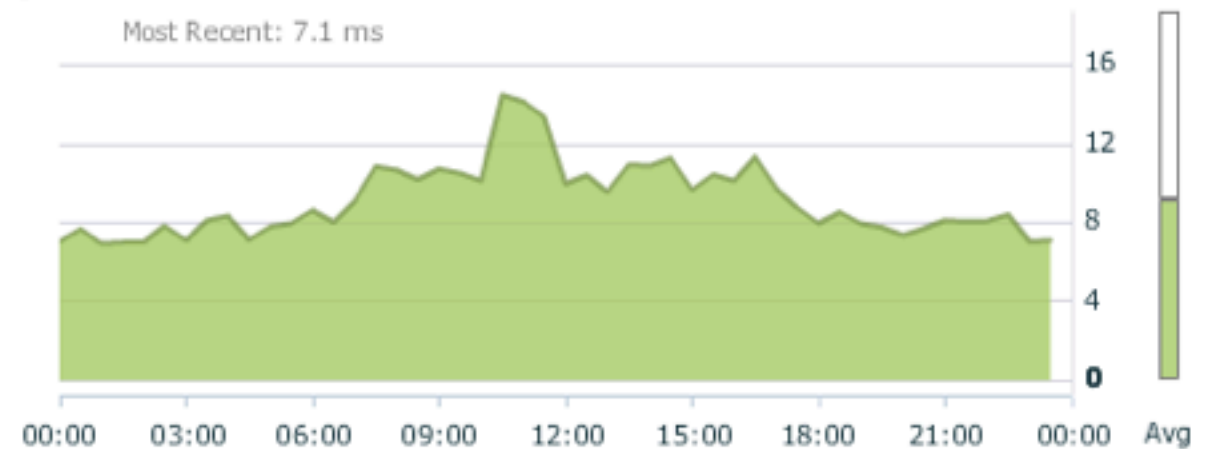
Feb 01, 2011 10:30 - 11:00

	Average	Min	Max
Voice Loss	0 %	0	0



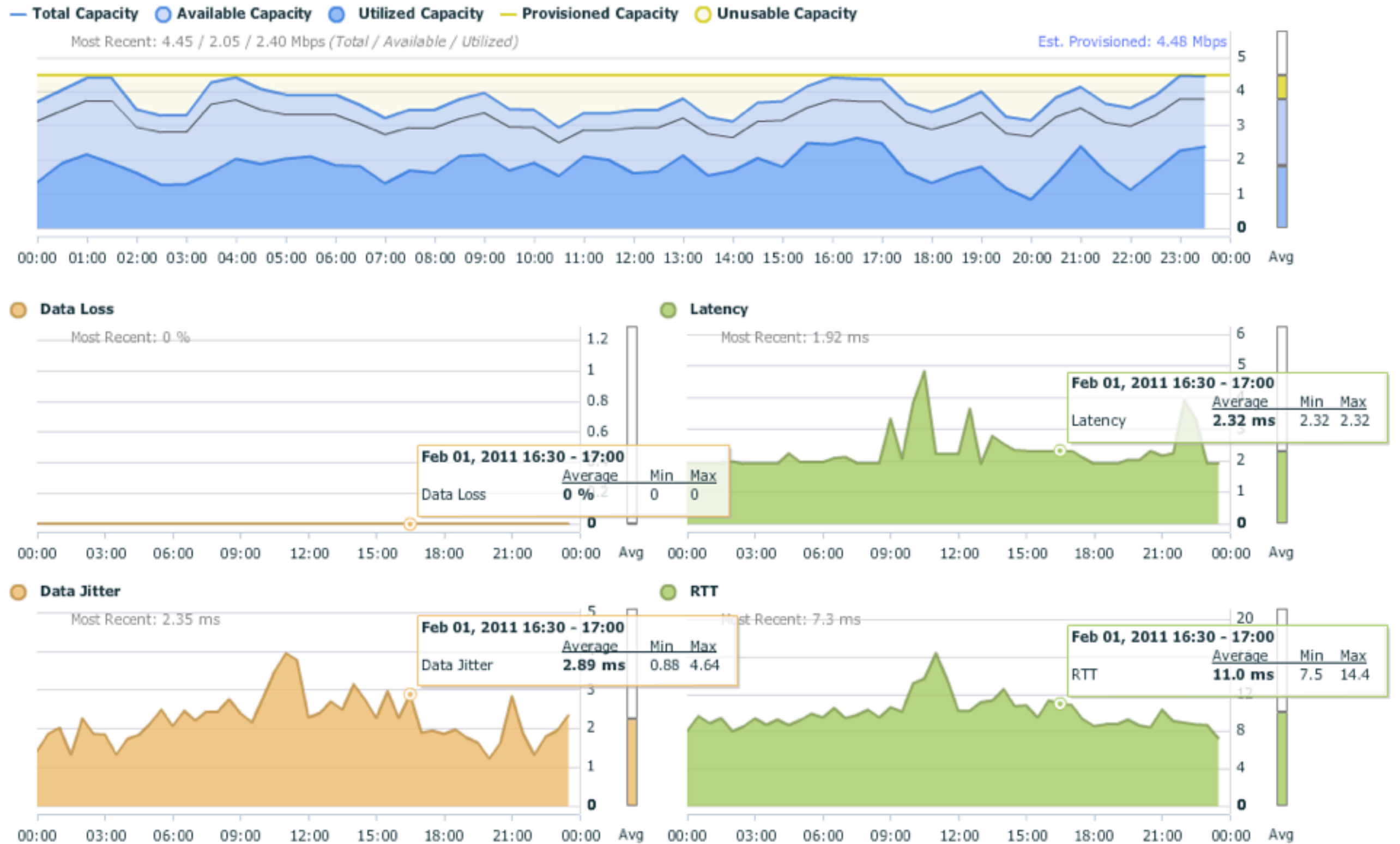
RTT

Most Recent: 7.1 ms



network assessment

unwired
revolution



network assessment

unwired

Summary Voice Detail


Hide Diagnostics

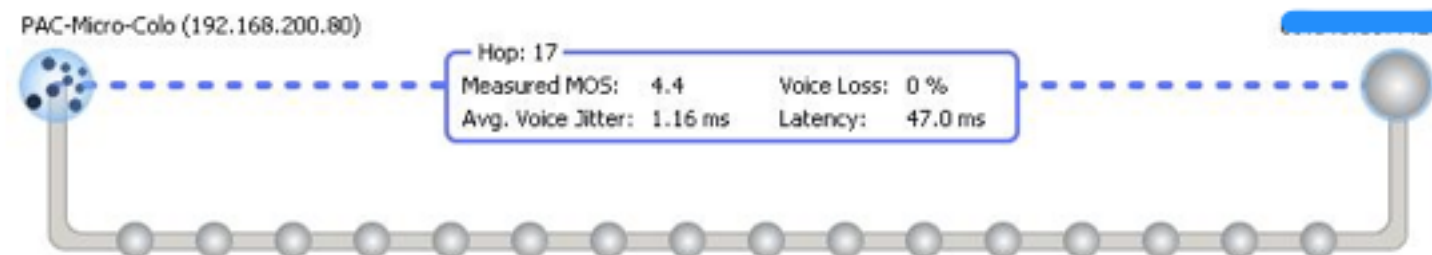
Hop	Severity	IP Address	Host Name	Voice Loss (%)	Measured MOS	Best MOS	Worst MOS	Latency (ms)	Voice Jitter (Avg/Max, ms)	RTT (Min/Avg/Max, ms)	QoS (Set/Measur)
1	✓	192.168.200.1	192.168.200.1	0	4.4	4.4	4.4	0.31	0.20 / 4.51	0.63 / 0.80 / 1.40	
2	!		.ip.mcleodusa.net	0	4.3	4.3	3.8	0.25	3.63 / 106	0.50 / 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.tengigabitethernet3-4.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	i	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								
		i 24%	Packet reordering detected [Relatively low levels should not affect network performance]								
15	i	68.86.210.154	te-1-1-ur01.middletown.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								
		i 32%	Packet reordering detected [Relatively low levels should not affect network performance]								
16	i	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								
		i 28%	Packet reordering detected [Relatively low levels should not affect network performance]								
17	i	69.141.187.42		0	4.4	4.4	4.4	47.0	1.16 / 12.3	94.1 / 97.0 / 130	- / -
		Frequency	Observation								
		i 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



five evolved pbx tips

4 ► 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



five evolved pbx tips

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