IT Expo East

January 24, 2007 Chris Sibley





Introduction to Cbeyond

- Cbeyond (NASDAQ: CBEY) is a Managed Service Provider (MSP)
- VoIP platform delivers hosted communications and data applications to small and medium enterprises (SMEs) on a monthly recurring basis

Markets:

- Atlanta
- Dallas-Fort Worth
- Denver
- Houston
- Chicago
- Los Angeles
- San Diego

Mission:

■ To give "big business" communications services to small businesses at prices they can afford – and that we can deliver profitably.



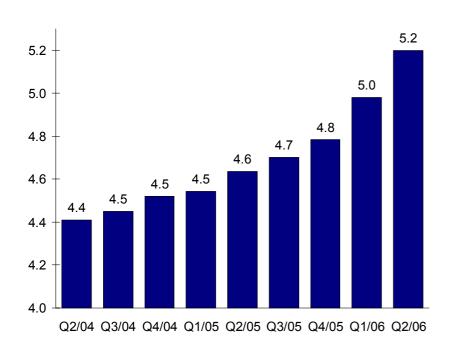
Introduction to Cbeyond

- Cbeyond's goal is provide an integrated suite of services that enable
 SMEs to effectively compete with larger enterprises
- Voice is just one of the applications that we can "bundle in" the communications package
 - But we do offer the complete range of voice services, including:
 - Local / LD termination
 - Voicemail
 - A dedicated toll-free number for every customer
 - Network-based conference calling
 - Calling cards



Introduction to Cbeyond

Applications Used per Customer



Application Portfolio

Core Services

- Local, Long Distance and Broadband Internet
- CbeyondOnlineTM a Web portal for customer self-care
- Voice interfaces: Analog, Digital (CAS/PRI) and SIPconnect

Mobile

Remote Access

Messaging

- Voice & Data
 - LG[™] flip-phone
- BlackBerryTM
- Email
- SMS

- VPN Site-to-Site
- VPN Remote User
- BeyondOffice
- Calling Card
- 「■ Voicemail
- Email
- SPAMblocker
- Whalemail
- Fax-to-email

Collaboration

- Filesharing
- Conference Calling

Commerce

- Web Hosting
- Web Kiosk
- Toll Free

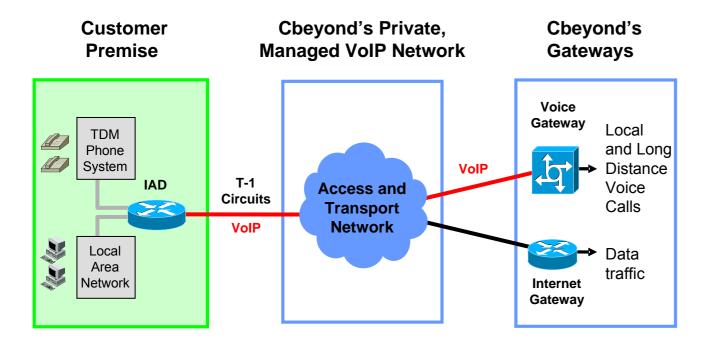
Security

- Managed Firewall
- Virus filtering
- Computer Backup

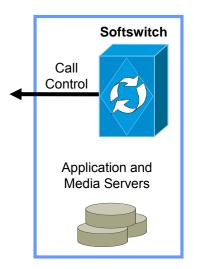
Cbeyond enables small businesses to become more efficient and project a larger image.



VoIP Integrated Access – "The Cbeyond Model"



Cbeyond's Data Center



- The Integrated Access Device (IAD) connects to customer's <u>existing</u> <u>phone system</u> and LAN
- IAD converts traditional voice calls into packets using VoIP
- To ensure quality of service, voice packets receive priority over data packets and are routed over reliable T-1 circuits
- Local calls are delivered to the ILEC and long distance calls are delivered to IXC partners
- Data traffic is routed to the Public Internet
- The IAD is controlled by Cbeyond's softswitch, which routes calls to the correct Voice Gateways



Benefits of VoIP Integrated Access

 SMEs can benefit today from integrated bundles such as those offered by Cbeyond, without making any new capital investments on PBX systems / equipment



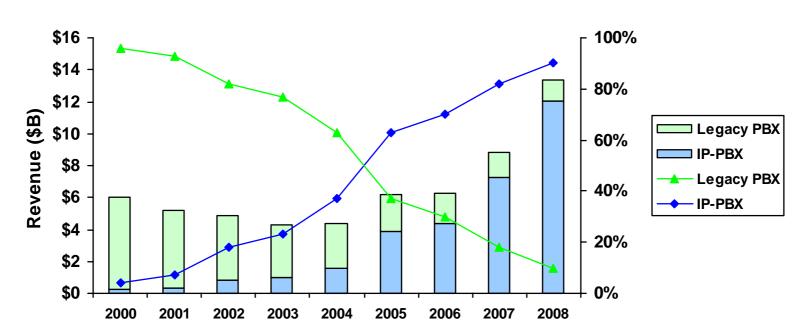
 An integrated bundle based on VoIP technology includes features not available with traditional TDM solutions

- Can utilize the entire access circuit for Internet access when no phone calls are active
- Access to a web-based customer portal that gives you complete control over your phone features
- Can use more than one T1 for access, and if one fails, both voice and data services will survive.

IP-PBX Revenue Forecast

... IP PBXs make up 80% of PBX lines shipped ...by 2008, 55% of all <u>deployed</u> PBXs are expected be VoIP enabled

U.S. PBX Revenue Forecast



Sources: Synergy Research, Frost & Sullivan, CompTIA



IP-PBX: Market Trends

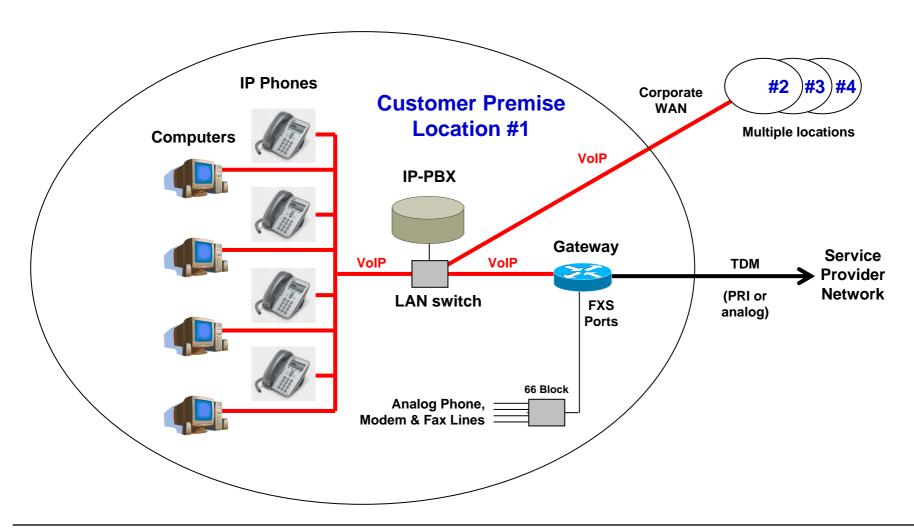
 Rapidly evolving IP-PBX market allows small and medium enterprises to take advantage of features typically available only to large enterprises

Key Market Trends:

- Direct Inward Dial (i.e., individual phone numbers for all employees) serves as a primary driver for small business users
- Desktop and application integration (unified messaging, corporate directories, instant messaging, presence management, collaboration, contact centers, CRM integration)
- Simplified Moves, Adds and Changes (MACs) and system management
- Multi-location business solutions (i.e. single PBX servicing multiple locations)
- Geographic independence (telecommuters and road warriors)
- Soft-phone support
- Single physical wiring infrastructure
- "PBX" is becoming a software application
- Multiple packages with 10 phones now less than \$4000

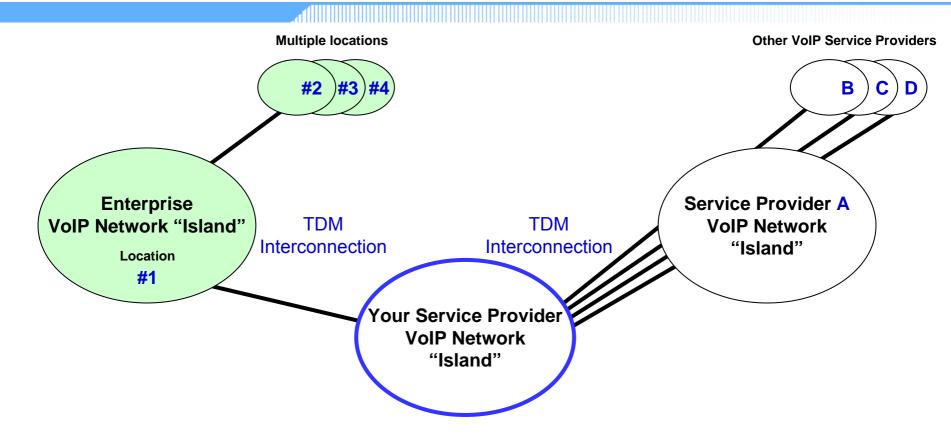


"Traditional" IP-PBX Connectivity Overview





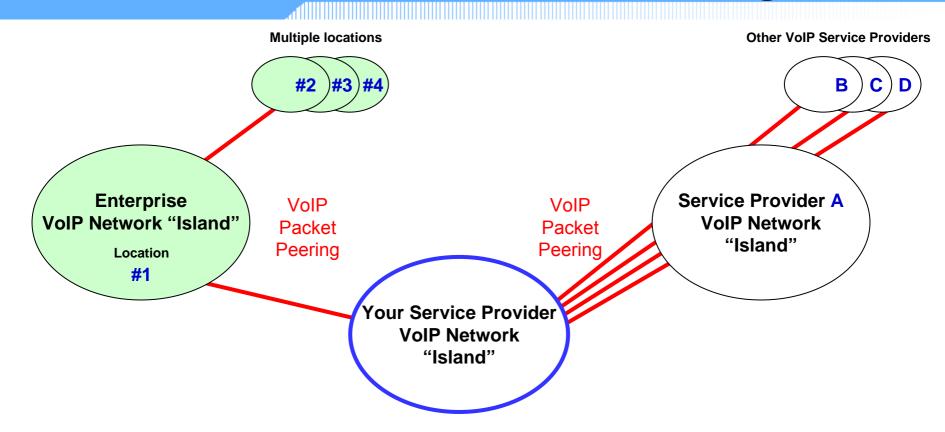
TDM Connections Produce Isolated VoIP Islands



Today, most VoIP networks connect to other VoIP networks by converting from VoIP ("new world") back to TDM ("old world") and then back to VoIP ("new world").



The Next Generation – Direct IP Peering



- Solution As more unique VolP Networks are created, the demand for VolP packet peering increases
- Eliminating gateways reduces the investment required and increases quality by reducing latency
- Inter-domain support for "IP Only" features
- Session Initiation Protocol (SIP) helps to unify the various VoIP protocols (H.323, SCCP, MGCP, SS7)



SIP connect - The Next Generation, Today

- A Standards-based Approach for Direct IP Interoperability between IP PBXs and VoIP Service Provider Networks
- SIPconnect meets a logical requirement for interfacing IP PBXs with service provider networks.
- SIPconnect was developed with broad input from leading companies and SIP luminaries from the IETF.
 - The SIPconnect Interface Specification was launched by Cbeyond Communications in 2005 with support from Avaya, BroadSoft, Centrepoint Technologies, Cisco Systems, and Mitel. Subsequent to an initial draft released by this group, the SIPconnect group and the SIP Forum worked to transfer this activity into the SIP Forum's Technical Working Group for further development and formalization into the current SIP Forum Recommendation.
 - To download a copy of the specification, navigate to www.sipforum.org/sipconnect

SIP connect - Key Benefits

A Ubiquitous Approach

 SIPconnect provides a common method for IP peering between SIP-enabled IP PBXs and VoIP service providers

Standards Based

 SIPconnect leverages existing SIP and related VoIP standards published by the Internet Engineering Task Force (IETF)

Customer Cost Savings

 Peering will lower service provider infrastructure cost and reduce the need for customer premises gateways

Richer Feature Support

 SIPconnect improves the "communications experience" by allowing service providers to deliver enhanced, personalized services to IP-PBXs (and its <u>end-users</u>) and extends rich-media services enabled by IP-PBXs across service provider networks

Quality of Service

 Methods for handling QoS configuration, echo cancellation, DTMF relay, packetization rates, codec support and fax and data traffic are defined



SIP connect – Why is SIP connect Needed?

- Prior to SIPconnect, the state of standards for connecting an IP PBX with a VoIP service provider were largely undefined and predictable interoperability did not exist.
- "SIP" alone does not mean compatibility. "SIP" is comprised of 45+ RFCs, 25+ related RFCs (e.g. SDP), and numerous Internet Drafts. Prior to SIPconnect, the specific support required by an IP PBX and service provider was loosely defined as it concerns IP PBX to service provider peering.
- This lack of clear definition creates an environment where you can not take for granted that an IP PBX will work correctly when connected with a service provider network just because they "support SIP."
- Service providers and PBX manufacturers were confined to endorsing only tested and certified combinations of equipment and service offerings.



SIP connect – What Does it Really Define?

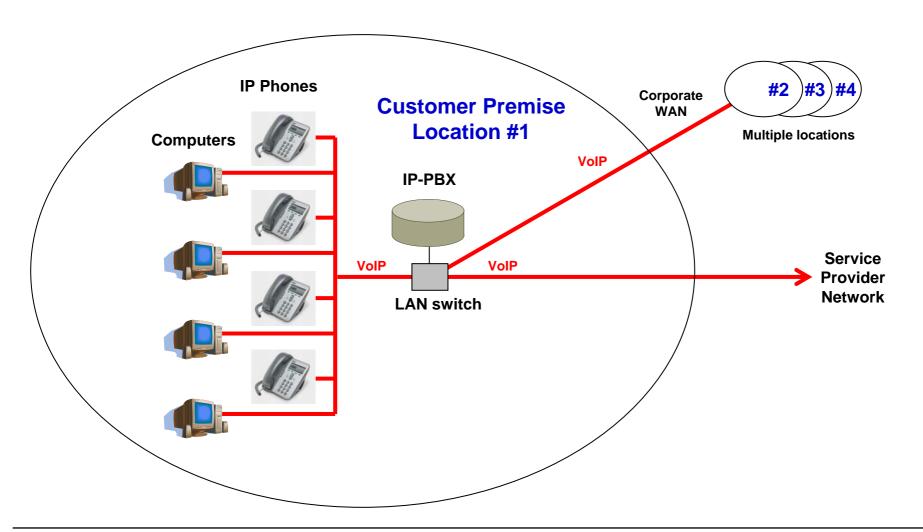
- SIPconnect is a standards-based method of interconnection between IP PBXs and VoIP service provider networks. It specifies a reference architecture, required protocols and features, and implementation rules necessary for seamless peering between IP PBXs and VoIP service providers.
 - Specifies the minimum set of SIP RFCs that must be supported by an IP PBX and service provider for the purpose of SIP trunking.
 - Provides precise methods for conveying user identity, including scenarios that involve the use of E.164 phone numbers
 - Ensures proper service security by clarifying the use of TLS for service authentication.
 - Clarifies minimum media requirements for issues such as codec selection, echo cancellation, fax and modem support, etc.

SIP connect - Who Supports It?

- Allworx
- Asterisk Open Source PBX
- Avaya
- Cbeyond
- Cisco Systems
- Epygi
- Fonality
- FrameWRx
- Linksys
- NEC
- Siemens
- Sphere
- Switchvox
- TalkSwitch
- Telechoice
- Vertical Televantage

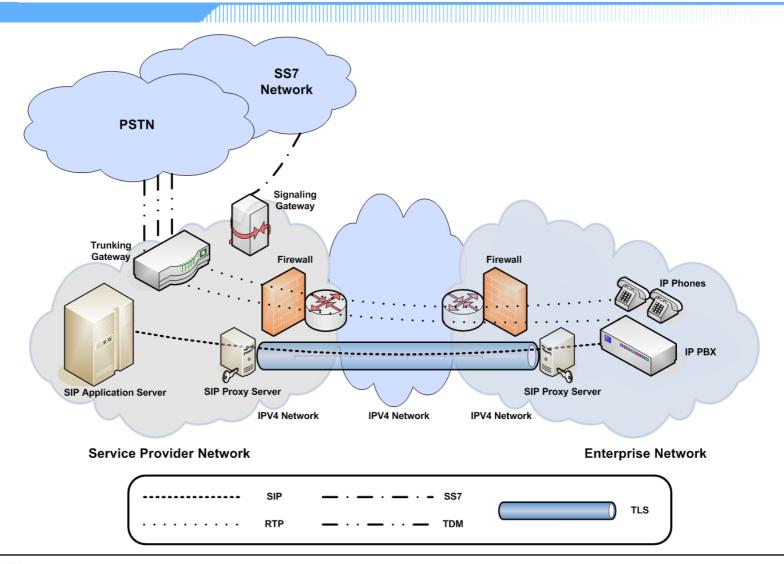


IP-PBX Connectivity Overview with SIP connect



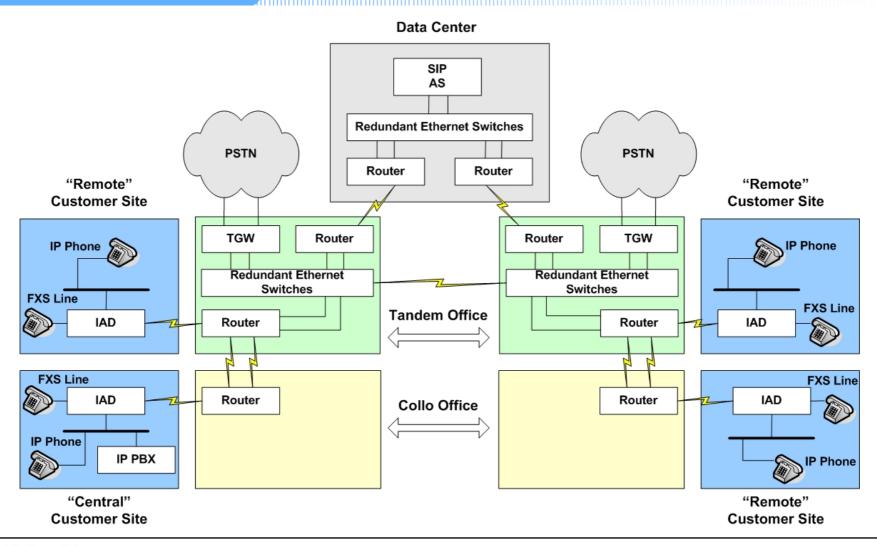


The SIP connect Reference Architecture





Cbeyond's Multi-Site IP PBX Architecture





Cbeyond's Multi-Site IP PBX Architecture

• Assumes Central and Remote sites are on-net

- Guaranteed QOS if all devices mark packets as defined in the SIPconnect specification
- Could easily support off-net sites, but there are some practical issues (e.g. 911 call routing, can't guarantee QOS over the open Internet)

Phones at the Remote Sites register up to the PBX at the Central Site

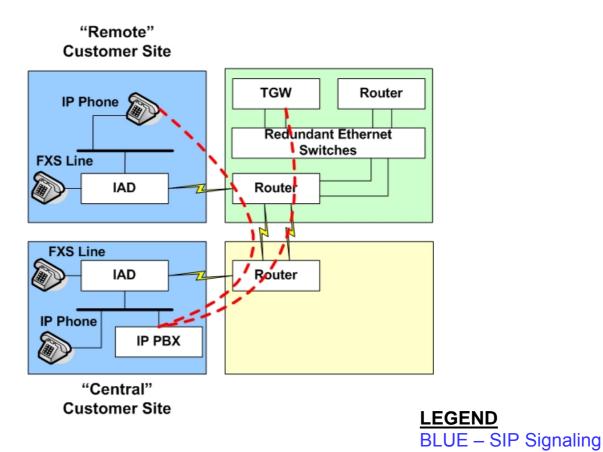
 The IAD and/or firewall/SBC at the Remote Sites are responsible for any necessary NAT "fix-up" functions

All calls to the Central and Remote Sites are sent to the PBX at the Central Site

- The IAD and/or firewall/SBC at the Central Site is responsible for any necessary NAT "fix-up" functions
- Media path is initially "hairpinned" through the PBX
- PBX can "release" the media to flow directly between the endpoints using the SIP RE-INVITE method



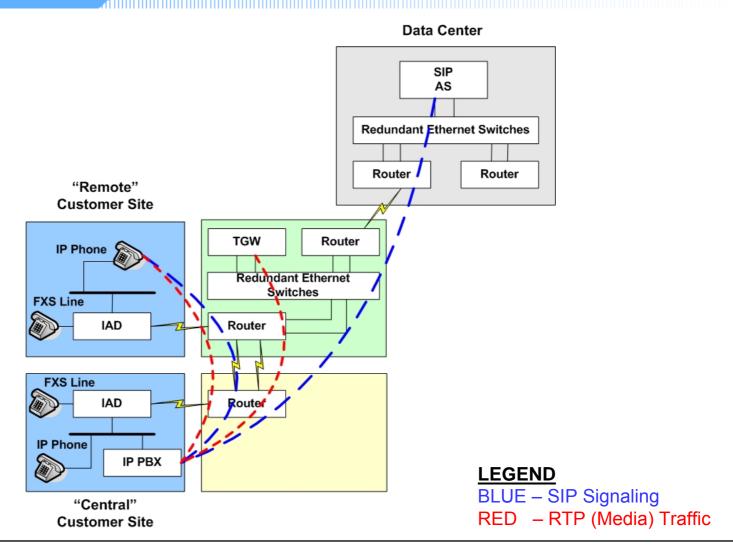
"Hairpinning" Media Through the IP PBX





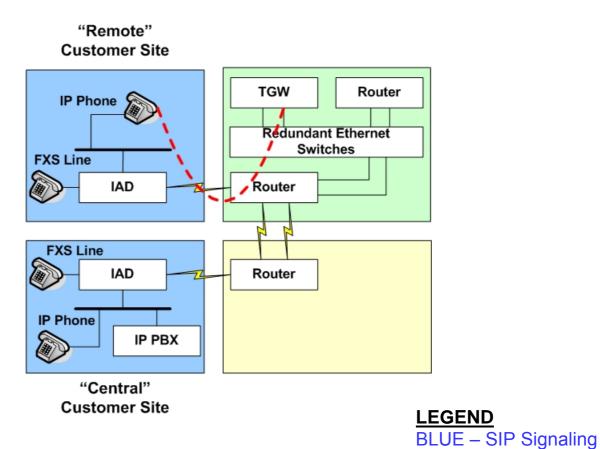
RED - RTP (Media) Traffic

Using RE-INVITE to "Release" the Media





Media Path Post-RE-INVITE





RED - RTP (Media) Traffic

Summary

- VoIP is transforming communications, and is enabling new business concepts like Cbeyond
- Short term opportunity for the SME includes higher value services from Managed Service Providers and BYOB Telephony Providers
- VoIP technology allows SMEs to more effectively compete with their larger peers
- It's time to throw out the old key system and buy a new IP PBX or hosted alternative
- Direct IP peering will enable a world of personalized communications as well as drive further cost savings





The Last Communications Company
A Small Business Will Ever Need.SM

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