

# Official Show Guide

## Diamond Sponsors:



## Platinum Sponsors:



## Featuring:



## Platinum Sponsors:



**WIN**  
 A Brand New  
 Toyota Prius!  
 Details on page 72





# Our communication technology is optimized for your sector... **ARE YOU?**

## 4 reasons why we're number one for communications technologies

### Contact center

Contact centers are all about people, processes and technology – and Aculab provides key enabling technologies to help you succeed with your intelligent customer contact solutions. Reliable, scalable products underpin virtual, hosted and networked operations across present and next generation infrastructures. Our market proven media processing resources and connection options are used worldwide in mission-critical, inbound and outbound customer interaction centres, helping to transform the way businesses communicate.

### Conferencing

Global constituencies need to communicate and converged collaboration services are increasingly used to unite geographically diverse audiences. If your business is helping others improve their internal and external communications through multimedia, toll-free, reservation-less or unified conferencing, Aculab offers state-of-the-art enabling technologies to help you. Our fully featured, high density, low cost per channel and superior voice quality proposition is ideal for carrier-grade audio and VoIP-based conferencing platforms.

### Fax technology

If your primary business is fax – whether outsourcing or hosting fax services or deploying fax servers – Aculab offers the reliability, scalability, and compliance you need to succeed. As the market leader in fax enabling technology, Aculab should be at the heart of your enterprise-class, hard-copy data management and delivery solutions. Our market proven products can help you deliver mission critical desktop, broadcast or production-class fax in many business sectors.

### Military applications

Aculab has become a centre of excellence for many agencies delivering network centric communications solutions for deployment in tactical and infrastructure networks throughout the defense technology sector. Our advanced enabling technologies and responsive commitment to bespoke engineering development provide essential interoperability and highly effective, redundant, reliable and scalable components that enable our partners to deliver secure applications for joint land, sea and air operations.

**Booth 314**

## Our difference

When you invest in a partnership with Aculab, you get more than the highest quality media processing technology that has been optimized for your sector. Whether it's for contact centers, conferencing, fax technology or military applications, you get the reassurance of a stable and proactive partner, offering continued support and enabling you to adapt to change before it happens.

## Our technology brings you

- Reduced operating costs
- Faster time to market
- Improved margins and ROI

### For further information...

**[www.aculab.com/4reasons](http://www.aculab.com/4reasons)**  
**[info@aculab.com](mailto:info@aculab.com)**  
**+1 781 433 6000**  
**+44 (0) 1908 273802**



The technology **driving** your success





Dear Colleague,

It is my great pleasure to welcome you to the 18th INTERNET TELEPHONY® Conference & EXPO. At TMC, we are very proud to have been hosting this event since 1999 and believe this is the best one yet. Thank you for joining us.

There are so many great events taking place over the three days of the conference, please take a few minutes to review the highlights below, and use this directory as your guide to make sure you don't miss a thing. For times and locations, please see the Agenda-At-A-Glance on pages 42 & 43.

## Tuesday September 16:

- Conference breakout sessions begin at 9:00am \* (Page 42-43)
- TMC University Certification programs: Open Source\*; Microsoft OCS\* (sponsored by Quintum); and FMC/Mobility\*; SIP in the Contact Center (Page 10)
- InGate's FREE SIP Trunking Workshop (Page 16)
- Avaya keynote at collocated Communications Developer Conference (Page 22)
- Skype, Microsoft, and Sonus Networks keynote speeches begin at 3:30pm (Page 22)
- Fonality's trixbox Training Courses\* (Page 12)
- Welcome Networking Reception at 5:00pm, hosted by Dash Carrier Service (Page 22)

## Wednesday September 17:

- Conference breakout sessions begin at 9:00am \* (Page 42-43)
- TMC University Certification programs: IP Network Certification\*; SaaS Certification (sponsored by Intelpeer)\* (Page 10)
- FREE Reseller Solutions Day workshop, sponsored by Interactive Intelligence (Page 18)
- Fonality's trixbox Training Courses\* (Page 12)
- Ingate's FREE SIP Trunking Workshop (Page 16)
- Cisco and Texas Instruments keynotes at collocated Communications Developer Conference (Page 22)
- AudioCodes, BroadSoft and 8x8, Inc. keynote speeches begin at 2:30 pm (Page 22)
- SIP Forum SIPConnect Workshop (Page 16)
- Exhibits Open 4:00 pm - 8:00 pm
- Aculab hosts the Exhibit Hall Grand Opening Networking Reception beginning at 6:00 pm.

## Thursday September 18:

- Conference breakout sessions begin at 8:15am \* (Page 42-43)
- TMC University Certification programs: Unified Communications \*; SIP: The Foundation for Building Enhanced Services, sponsored by Dialogic \*; Next Generation Call Center Management Certification\* (Page 10)
- Fonality's trixbox Training Courses\* (Page 12)
- Voice Peering Fabric FREE Workshop (Page 18)
- Exhibit Hall Open 11:00am - 5:00pm
- Giveaway of TOYOTA PRIUS HYBRID at 4:55pm! (Page 72)

\*Conference fee required for admission. All other events free for all badge holders.

Once again, I can't thank you enough for helping make this our biggest and best ITEXPO yet. And as always, please let me or a member of the TMC staff know what you think of the event. You'll find us walking about the conference all week, or you can stop by the registration area to give us your feedback. Have a great show.

Rich Tehrani

TMC President, Group Editor-in-Chief, and Conference Chairman

## Table Of Contents

Floor Plan .....	6
Exhibitor List.....	7
TMC University .....	10
trixbox Training .....	12
FREE Workshops.....	16-18
Keynote Presentations .....	22
<b>Call Center 2.0</b>	
Exhibitor Descriptions .....	24-30
<b>Internet Telephony</b>	
Exhibitor Descriptions .....	32
Agenda-At-A-Glance .....	42-43
Toyota Prius Giveaway Details ....	72
Q & A With Aculab.....	76



## Diamond Sponsors:



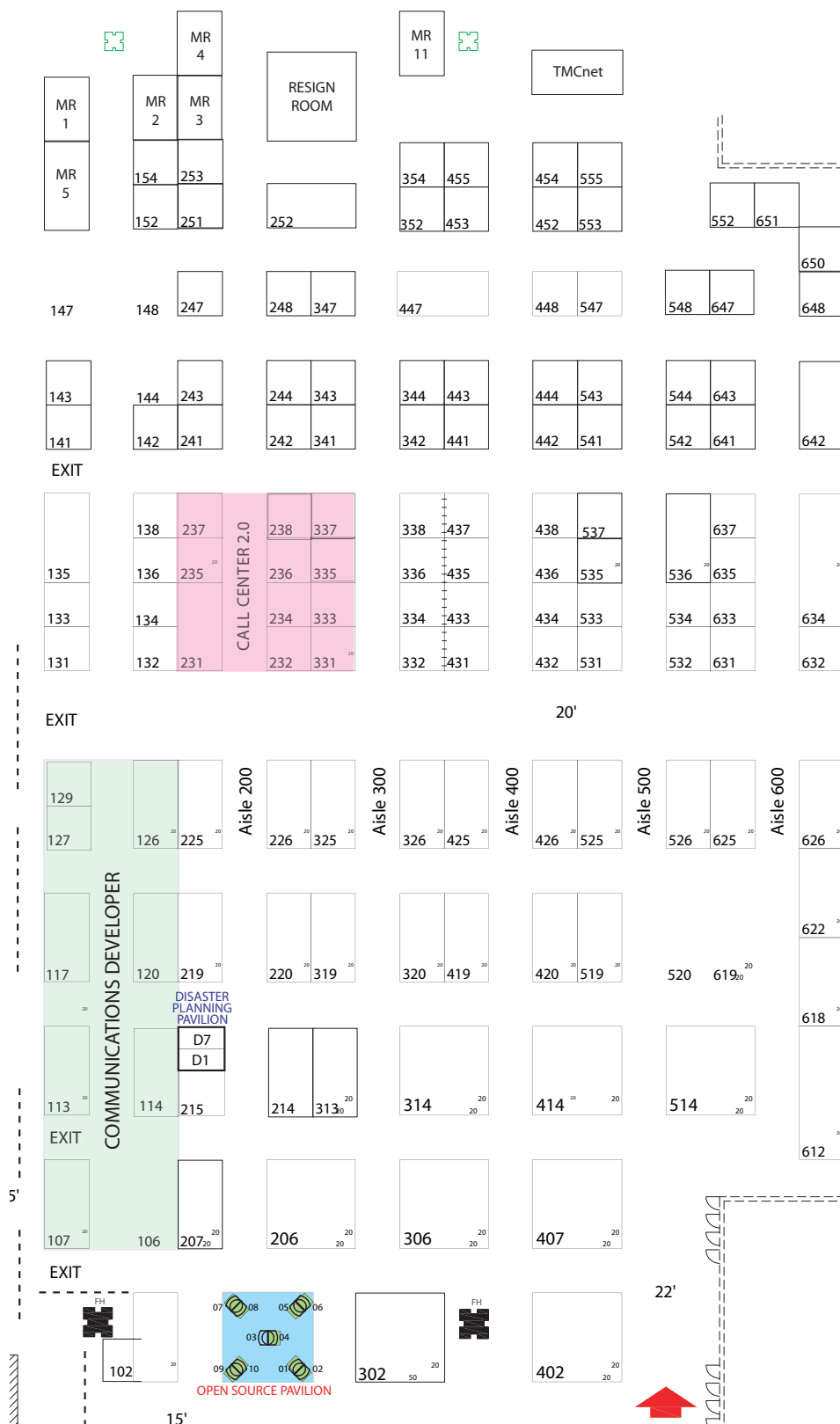
## Platinum Sponsors:



## Gold Sponsors:









# Exhibitor List



(Call Center 2.0 = Red • Open Source Pavillion = Blue • Communications Developer = Green)

360networks .....	Meeting Room 501B	Microsoft .....	Meeting Room 502B1
8x8, Inc. ....	407	Minuteman/Para Systems, Inc. ....	332
Aastra Telecom .....	OS 07 & 08	Mitel .....	302
ABP Technology .....	618	MobileMax .....	436
Aculab .....	314	Multi-Tech Systems, Inc. ....	632
ADTRAN, Inc. ....	612	NESSOCIET, INC. ....	132
AireSpring .....	525	NetSapiens Inc. ....	425
AiTech .....	537	NETXUSA, Inc. ....	206
Altai Technologies .....	618	NMS Communications .....	CommDev 106
APEX Voice Communications .....	531	OAISYS .....	CC 337
ApplianX .....	314	Octasic .....	Meeting Room 4
Aspect Software .....	CC 232	OpenVox Communication Co, Ltd .....	437
Asteria Solutions Group .....	OS 01	OrecX .....	OS 05
AudioCodes .....	320	Packet Island, Inc. ....	338
Avaya - DevConnect Program .....	ComDev 114	Pactolus Communications Software Corp. ....	319
Basis Audionet .....	313	PAETEC .....	441
BillCall - A Panamax Infotech Division .....	142	Pangea Communications Corp. ....	633
Business Solutions .....	253	PhoenixSoft, Inc. ....	342
CamrivoX .....	618	Phonefusion .....	244
Cbeyond .....	438	Phonevite .....	344
ChannelVision Magazine (Beka Publishing) .....	650	PIKA Technologies Inc. ....	ComDev 102
Cisco .....	ComDev 120	Plantronics .....	643
ClearOne .....	618	PortaOne, Inc. ....	431
Commetrex Corporation .....	ComDev 127	Pulse, Inc. ....	534
CosmoCom .....	CC 231	Quanta Syspine .....	548
Critical Links .....	432	Quantum Technologies and Network Equipment Technologies, Inc. ....	420
CyberData Corporation .....	131	Qwest Communications .....	426
Cylogistics .....	326	Redfone Communications LLC .....	OS 09
Dash Carrier Services .....	622	Ring Carrier .....	326
Dialexia Communicatians Inc. ....	433	RNK Communications .....	625
Dialogic .....	SPONSOR	SANGOMA .....	519
DIDX & Super Technologies .....	154	Sansay .....	220
DIGITALK Ltd .....	207	Seawolf Technologies, Inc. ....	631
Digium .....	OS 06	SimpleSignal .....	642
Eastwind Communications .....	ComDev 113	SIP PRINT LLC .....	444
emFAST Inc. ....	147	SiTel Semiconductor .....	Meeting Room
Epygi Technologies Ltd. ....	434	snom technology AG .....	651
Excendia .....	442	SoTel IP Services .....	133
FaxBack, Inc. ....	533	Spenser Communications, Inc. ....	544
FaxCore, Inc. ....	D1	Spracht .....	647
FaxSIPit Services Inc. ....	535	Stealth Communications .....	CC 335
Fonality, Inc. ....	402 & OS 02	Stratus Technologies .....	547
FreedomVOICE Systems .....	634	Syntellect .....	CC 333
Freeside Billing .....	OS 10	Synway .....	ComDev 117
G-Tek .....	336	Taqua LLC .....	637
Global IP Solutions (GIPS) .....	ComDev 129	Telcobridges .....	ComDev 619
GlobalThink Technology SA .....	341	TeleMatrix / Scitec .....	443
Gold Systems .....	148	Texas Instruments .....	ComDev
GotVoice, Inc. ....	541 & Meeting Room 1	Tigerpaw Software .....	648
Grande Communications .....	447	Time Warner Cable Business Class .....	414
Grandstream Networks .....	419	TouchStar .....	CC 331
Gridborg America .....	448	Touchstone Technologies, Inc. ....	526
i2Telecom .....	Meeting Room 2	tw telecom .....	552
IgeaCare Systems Inc. ....	D7	U4EA Technologies .....	542
iKnowWare .....	225	Vitelity Communications - Wholesale Carrier Services .....	219
iLinc .....	626	VIXXI Solutions, Inc. ....	536
Infineon Technologies .....	Meeting Room 5	VOCALCOM .....	226
Ingate Systems .....	Meeting Room 507	VoIP Logic .....	435
IntelePeer .....	135	VoIPConsultants.biz, LLC .....	343
Interactive Intelligence .....	514	VoIPstreet .....	243
Intuitive Voice Technology .....	325	Volt Delta Hosted Solutions .....	306
IR link corp. ....	252	Voxbone .....	334
Kentrox .....	248	Voxeo Corporation .....	ComDev 126
LAN Power Systems .....	326	WBS Connect .....	543
Macquarie Equipment Finance .....	144	Wyde Voice, LLC .....	532
MARSHALL ELECTRONICS .....	347	XO Communications, Inc. ....	635
Matrix Telecom Pvt Ltd .....	641	Xorcom .....	OS 04
MERA Systems .....	241	Zeacom Communications Center .....	520
Message Technologies, Inc. ....	242		





#### QUICK FACTS

- Over 100 offices worldwide
- Over 3,000 employees
- Operating in over 90 countries
- Over 1,500 resellers
- #1 in the U.S. IP SMB market (up to 500 employees) in 2006 (Source: "InfoTrack for Unified Communications": InfoTech, April 2007)
- #1 in the U.K. enterprise communications market (Source: MZA, IQ2007)
- #2 in Western Europe in terms of IP extensions shipped (Source: MZA, IQ2007)

## About Mitel – U.S. Region

Mitel® is a leading provider of voice, video, and collaborative communications solutions and services for businesses of all sizes. We operate in over 90 countries, offering easy access to our sales and service organization via our network of over 100 offices, which support over 1,500 value-added resellers and partners. Our entrepreneurial heritage, built upon agility and flexibility in the way we serve our customers, has continued over the company's history from the introduction of the first private branch exchange in 1978 to our many innovations in IP communications. Our position as an early innovator and industry consolidator has propelled us into our current market-leading position as the telecommunications industry shifts from legacy technology to IP-based systems.

### Mitel Solutions

Our solutions are designed to satisfy the needs of all types and sizes of businesses. With unmatched reliability and scalability, Mitel's solutions enhance employee productivity, increase customer service, reduce costs, and generate new revenue streams. Our approach is unique. We see communications as an investment in a business' ongoing performance—not a one-time equipment acquisition. Our focus on understanding a business' needs means we provide tailored industry specific and user-centric solutions maximizing our customers' return on their investments.

Our portfolio includes communications platforms and gateways, business telephones, and unified communications software applications. Whether for small offices or multinational enterprises, Mitel has the solutions, including hosted offerings and end-to-end managed services, to suit any business model. Our IP migration strategy enables businesses to maintain their existing infrastructure investment while taking advantage of the latest IP technologies. Leading the unified communications market with the latest in presence integration, we provide businesses with what they want—choice.





## Mitel Services Portfolio

Mitel offers a full spectrum of managed services solutions that effectively address business communications needs in the United States. Mitel DataNet delivers a single source for increasingly vital data services, ensuring that all data-related products are designed, configured, installed, and supported to provide optimum communications performance for the long term. Our NetSolutions® division provides a best-in-class, diverse portfolio of Tier 1 regional and national telecommunications services paired with unrivaled customer support. Mitel CommSource® further enhances communications solutions with over 4,000 products and support from many of the best vendors in the industry.

## Managed Services Program

Our TotalSolution Program addresses the total cost of ownership with end-to-end business communications services and solutions. Mitel managed services enable companies to focus on what they do best—running their business. Through relationships with carriers and preferred suppliers, we can take care of all of our customers communications needs for a simple, fixed monthly fee.

## Mitel's Portfolio

**Business Telephones** – Our phones are designed with the individual user in mind, while providing essential business collaboration tools with presence integration. Available in multi-button configurations or with intuitive graphical interfaces, Mitel phones can be industry-tailored to suit both standard and specialty applications—including industry-specific terminals for attendants, contact center agents, and command-and-control applications, as well as reliable, secure desktop devices for everyday office use.

**Communications Systems** – Reliable, flexible, and scalable, Mitel's communications systems, including Inter-Tel 3000, Inter-Tel 5000, Mitel SX-200 IP Communications Platform (ICP), and Mitel 3300 IP Communications Platform (ICP), offer market-leading choice, with best-of-breed options for the small office right up to the multinational enterprise.

**Collaboration** – Extend teamworking beyond office boundaries with efficient access to information and people, using leading-edge applications such as Mitel Quick Conference, Mitel Your Assistant™, and Inter-Tel Audio and Web Conferencing, and Inter-Tel Unified Communicator. Mitel's solutions offer relevant, presence-enabled business communications and collaboration capabilities to individuals inside and outside of the organization, independent of physical location.

**Contact Center** – Reduce costs, intelligently service contacts across multiple channels, and let agents focus on high-value interactions while increasing customer satisfaction. Mitel's customer interaction solutions allow the effective and efficient management of contact centers for both small and large applications and are designed with customer retention—and acquisition—in mind.

**Messaging** – From basic voice mail to advanced unified messaging, Mitel's Messaging Solutions, including embedded messaging and Mitel NuPoint Messenger™ IP, increase personal productivity, improve accessibility, and protect infrastructure investment.

**Mobility** – Successful decision making requires immediate access to the right people at the right time. Mitel offers solutions for personal, campus, or enterprise mobility—addressing the needs of the corridor warrior to road warrior. Mobile connectivity, utilizing Mitel's Mobile Extension, drives smart business, lowers costs, delivers a competitive advantage, and delivers results “in the moment”.

**Teleworking** – This plug-and-work solution, using a full range of desktop or soft phones, allows companies to offer workers-on-the-go with all the features and functionality of their office phone and full presence at work—wherever they are.

## Mitel Strategic Partners

Mitel and our partners develop integrated and innovative solutions, leveraging technology from industry leaders, such as Microsoft®, ProCurve Networking by HP, and Sun Microsystems™. Mitel makes it a priority to build long-term relationships with other companies to create better solutions for our customers. Mitel's Solutions Alliance provides a selection of third-party software and hardware options that integrate seamlessly with Mitel's portfolio.

## Doing Business with Mitel

With Mitel it's about the relationship. While addressing the global market, Mitel meets local needs through our branch offices and our network of over 1,500 trained and certified resellers worldwide. Mitel's branch offices and Mitel authorized PARTNERS provide unparalleled customer service and support. Mitel insists on delivering not only an exceptional communications solution tailored to each business, but the support and services that make an initial communications investment pay long-term returns.

[www.mitel.com](http://www.mitel.com)






## TMC UNIVERSITY

### What You'll Gain:


- Independent accreditation for completing each course.
- Independently certified evidence that you possess advanced competence using these important technologies.
- Impressive certification from a respected source on your resume.
- Credentials to land lucrative consulting/reseller opportunities.
- Immediately become the expert called upon to lead your company's strategies.
- Enhance your chances for a promotion.




### Tuesday, September 16, 2008

Room:	Time:	Course:	
512	9:00am	Microsoft OCS	Hosted by: 
513	9:00am	FMC/Mobility	
514	9:00am	Open Source	
511BC	9:00am	SIP in the Contact Center	

### Wednesday, September 17, 2008

Room:	Time:	Course:	
514	9:00am	IP Network Security	Hosted by: 
511BC	9:00am	SaaS	

### Thursday, September 18, 2008

Room:	Time:	Course:	
512	9:00am	Unified Communications	Hosted by: 
514	9:00am	SIP: The Foundation for Building Enhanced Services	
511BC	9:00am	Next-Generation Call Center Management	

### Visit The Open Source Pavilion on the ITEXPO Exhibit Floor

The Open Source Pavilion is a "must-see" destination for all attendees seeking more information on how Open Source communications solutions can benefit you.





**Booth  
135**

# **Transform** Your Applications With **Rich Media, Voice and Video**

**with IntellePeer, the leader in  
on-demand communications services**

**Find out more about innovative  
Communications-as-a-Service**

Wednesday September 17, 2008 — 9:00 - 9:45am

Software-as-a-Service Room: 511B/C

Wednesday September 17, 2008 — 11:00 - 11:45am

CRM Software-as-a-Service Room: 511B/C



[www.intelepeer.com](http://www.intelepeer.com)

**Booth 135**



## Room: 503



### **Administrator Course** **Fonality tribox** **Open Communication Certification**

Complete details and registration at [www.itexpo.com](http://www.itexpo.com)

FtOCC CE is a 2.5-day course designed to teach the basics of tribox@CE installation and administration. FtOCC CE is a great opportunity to roll up your sleeves and learn the ins and outs of tribox CE. This course is designed for those who wish to install tribox CE systems for their company or their clients, people who will have to maintain a tribox installation, or users who are relatively new to tribox and who want a better understanding of its capabilities. FtOCC CE focuses on the configuring and troubleshooting a tribox CE system. From a complete understanding of all of the user interface features, to how to order and configure, telephone circuits, and how to do basic troubleshooting. At the end of the course you will receive a FtOCC CE certification.

#### **Who Should Attend?**

- IT Engineers tasked with supporting or implementing an IP PBX.
- System integrators and VARs who want to sell IP PBX systems.
- Phone system resellers looking to move into the IP PBX market

Each student receives the tribox VMWare image and the VMWare Player to install on his/her laptop for course lab work. Note that the course does not presume that the students are full-time sys admins, however all are presumed to have some general computer experience.

**Materials:** Polycom SIP Telephone, FtOCC Workbook, Jump drive with tribox VMWare and FtOCC presentations, "FtOCC Certified" polo shirt, tribox T-Shirt and cap

#### **Course Objectives:**

- Become comfortable working within the Linux shell.
- Understand the various components of Asterisk.
- Be able to add, delete and alter user configurations.
- Be able to configure several common models of SIP phones.
- Be able to create IAX2 or SIP links to service providers.
- Be able to alter basic parameters for Zaptel devices (FXS and FXO).
- Be able to add DIDs and direct them to users.
- Be able to implement new features.
- Be able to build IVR menus with proper loop-prevention, error handling, etc.
- Be able to record prompts for IVR menus, audio text messages, etc.
- Understand and use the Asterisk ACD (queue/agent) system.
- Understand the issues with NAT and their impact on remote users.
- Understand the strategies which can eliminate these issues.
- Be able to check on system health from the shell/CLI.
- Understand and prevent security threats that impact a tribox system.
- Understand how call routing works in Asterisk and be able to implement a basic Least Cost Routing structure in the Dial Plan.
- Recognize the issues related to emergency handling (911/E-911) and how best to configure a system to securely handle such calls.

## Room: 518



### **Technician Course** **Fonality tribox** **Open Communication Certification**

Complete details and registration at [www.itexpo.com](http://www.itexpo.com)

FtOCC Technician is a three-day technical certification course designed to train resellers and consultants to support their clients running tribox CE, tribox Pro, and PBXtra systems. Taught by Fonality technical support instructors and designed for Linux system administrators, FtOCC Technician dives deep into platform and application installation, carrier setup and integration, network configuration, echo causes and remedies, and other common issues. Upon completion of this course, resellers automatically become Fonality Authorized (AL2) Resellers and earn increased discounts on support.

#### **During the course, we will cover:**

- Tier 1 and Tier 2 level troubleshooting
- LAN & WAN environments
- Proper system hardware configuration
- tribox and PBXtra installation guidelines
- IP phone and SIP provisioning
- Proper router and firewall configuration

#### **Who Should Attend?**

- Resellers/consultants growing tribox and PBXtra businesses
- Linux technicians and engineers
- Fonality Authorized and Premium resellers

#### **Course Objectives:**

- At the end of the class you should:
- Install and configure tribox CE, tribox Pro, and PBXtra.
- Quickly resolve installation, administration, and usage problems.
- Provision your own phones, thereby decreasing the turnaround time required for add-on phones and devices!
- Experience fewer problems by avoiding the usual "beginner" mistakes!
- Have an in-depth understanding of how tribox and PBXtra work (product, process, policy, protocol).
- Receive full reseller discounts (for signed Authorized and Premium resellers only)
- Receive direct priority Level 2 access to tribox and PBXtra support services (for signed Authorized and Premium resellers).

#### **What do you get?**

- Free tribox Pro demo kit with Call Center Edition and support
- 10% discount on tribox Pro licenses
- 25% discount on support
- FtOCC Certified polo, t-shirt, and hat

*NOTE: A practitioner's knowledge of Linux is required. Without it, you will likely not gain much value from the class, as we start day one from the Linux command line interface and build from there. In particular, the hands-on laboratory exercises will require a comfort with basic CLI commands, simple scripting, and the ability to quickly scan and analyze system and application logs.*



Distributed by **cyLogistics™**



**When Cash Flow Really Matters**



**Class-5 Solutions on Open Source**



**Power over Ethernet (PoE) Products**



**TDM & T1 Controllers**



**Audio Conferencing solutions**

**See us at booth #326**



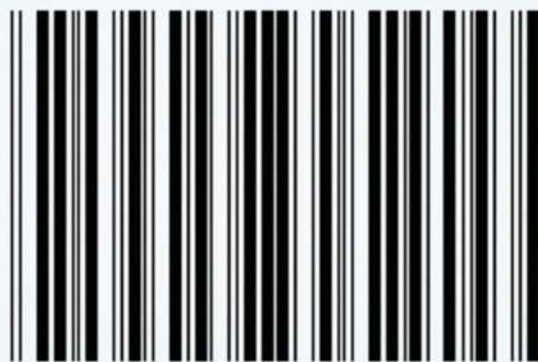
4920 El Camino Real, #108  
Los Altos, CA 94022  
<http://www.cylogistics.com>

650-694-4949  
Fax: 650-694-4953  
[sales@cylogistics.com](mailto:sales@cylogistics.com)





I am not a



024-8051



# **You have a unique business. Why not have a unique approach to your communications?**

When your business is treated like any other business, that can't be good for business. At Time Warner Cable Business Class we'll treat your company like the individual business it is, with a unique approach to business communications. You get a local, dedicated representative to help determine your needs, tailor a solution for your business and provide ongoing support when you need it. With our high-capacity network, we offer reliable Internet access (cable broadband or fiber optics), Business Class Phone, and Video, to help you stay productive. And make sure you get exactly what you need, without paying for what you don't.

## **Visit us at Booth 414**

**1-866-TWC-4BIZ**  
[www.twcbc.com/la](http://www.twcbc.com/la)



Internet. Phone. Video.



## INGATE'S SIP TRUNKING WORKSHOP - ROOM 507

### SIP Trunking: Everything You Need to Know

Educating end users, resellers, distributors and systems integrators about SIP trunking is on the agenda as Ingate® Systems partners with Avaya, TMC, leading IP-PBX vendors, SIP trunking service providers and industry thought leaders to offer "SIP Trunking: Everything You Need to Know" seminars at the INTERNET TELEPHONY Conference & Expo West 2008.

These seminars will provide a comprehensive overview of SIP trunking, and include general information panels and technical insight sessions from both the service provider and enterprise perspectives.

#### New this season:

Step-by-step, interactive whiteboarding sessions with Avaya to illustrate the what, why and how of SIP trunking architecture

Two case study seminars - one featuring a SIP trunk installation for American Idol's production company, the other on bundled SIP trunk solutions - to delve deeper into the issues

SIP Trunking Boot Camp for "Basic Training" on SIP trunk installations with Avaya and Ingate

Security sessions with VOIPSA will discuss VoIP and SIP trunk security.

LIVE DEMOS of participants setting up a secure SIP trunk live, on-site will be featured, to showcase how easy it can be to deploy SIP trunks.

Attendees can earn a SIP Trunking Professional Certificate by participating in the Professional Development Program on the first day of the Show.

#### You'll learn:

- SIP trunking: benefits, issues, and opportunities
- How SIP trunking can maximize the Return on Investment for your IP PBX
- An installation roadmap, solutions for interoperability, etc.
- An introduction to SIP Trunking service providers, their plans, and requirements
- A complete tool kit to ease the installation of SIP trunks



SIP trunks reduce costs, as they eliminate the need to purchase local PSTN gateways, costly ISDN BRIs (Basic Rate Interfaces) or PRI's (Primary Rate Interfaces).

Join us for these informative seminars. FREE for all attendees.

#### Tuesday, September 16 SIP Trunking Professional Development Day

9:00 - 10:00 am:

**Introduction to SIP Trunking**

10:00 - 10:30 pm:

**LIVE DEMO: Setting Up a Secure SIP Trunk**

11:00 - 12:30 pm:

**The Enterprise Infrastructure**

1:30 - 3:00 pm:

**The Service Provider Perspective**

*A Professional Development Certificate will be awarded to those who attend all three Tuesday seminars.*

#### Wednesday, September 17 The Mechanics of SIP Trunking: Implementation, Issues, Solutions

8:30 - 9:45 am:

**Case Study: Benefits of Bundled SIP Trunk Solutions**

10:15 - 11:15 am:

**SIP Trunking & Security in an Enterprise Network**

1:15 - 2:30 pm:

**Case Study: American Idol Gets In Tune with SIP Trunking**

3:00 - 6:00 pm:

**SIP Forum SIPconnect Compliance Workshop**

### SIP FORUM

#### Thursday, September 18 SIP TRUNKING BOOT CAMP

8:30 - 10:30 am:

**SIP Trunk Basic Training with Avaya and Ingate**

#### Supported By:





# Accelerating the Development of **Intelligent Communications**

Competing in today's rapid-paced world requires the flexibility to adapt and innovate across all aspects of a business, from employee productivity and collaboration improvements, to more powerful and positive customer experiences. Through the Avaya DevConnect Developer and Partner program, Avaya gives you the resources, tools, education and support necessary to profoundly change the way businesses operate, turning communications in to a competitive advantage.

Your **free DevConnect membership** offers access to:

- Avaya SDKs, APIs and technical documentation
- No-charge remote labs and other developer tools
- Educational programs and sample applications
- One of the largest developer communities specifically focused on communications
- Opportunities for compliance testing & co-marketing.

© 2008 Avaya Inc. All Rights Reserved.

Visit **Booth # 114** to learn more, or join us today at **[www.avaya.com/DevConnect](http://www.avaya.com/DevConnect)**

# AVAYA

**INTELLIGENT COMMUNICATIONS**





## RESELLER SOLUTIONS DAY - ROOM: 502A

**Wednesday, September 17 - FREE Workshop for Resellers and Agents**

**Presented by:**



9:00 - 9:45 am

### **Making Money in IP Communications**

*Rich Tehrani, President, TMC*

*Don Gant, VP Bus Dev & Channel Mktg, Iwatsu Voice Networks*

*Robert Messer, President & CEO, ABP Technology, Inc.*

*Brit Vickner, Strategic Bus Dev Manager, Interactive Intelligence*

The communications market is filled with opportunity and pitfalls and navigating the field is something which requires experience and collaboration with your peers. In this session, expect an in-depth discussion of everything you need to become successful when selling communications products and services. We will explore unified communications, open source, FMC, WiFi telephony, mashups and CEBP. If you are unfamiliar with any of these acronyms (even if you are) you need to be at this free conference so you can learn what is happening, what is coming and how to benefit from it all.

The moderator of this panel will be Rich Tehrani, ITEXPO conference chairman, TMC president and the communications expert who has helped tens of thousands of resellers worldwide over the years sell PBX, ACD, call center, CTI and VoIP products and services.

This session is a must-attend for anyone and everyone involved in any way with communications sales to SMB or enterprise customers. This includes classic interconnects, data VARs, VADs, security resellers, etc.

10:00 - 10:45 am

### **Say Yes! To Unified Communications from the Beginning**

*Brit Vickner, Strategic Bus Dev Manager, Interactive Intelligence*

Learn why you should be selling and making more money with the right solution. ININ's Software Strategy that has proven to be a bet-

ter Approach to Complexity. An All-In-One open standard solution that provides the back office architecture (fabric) required for mass adoption. The options for Customers with a migration strategy (work with) or Greenfield (work without) requirements. Desktop productivity and User adoption is also a crucial component of a successful UC strategy with Multi-Media interactions no matter the device, place or user type when you need it.

12:45 - 1:30 pm

### **AXIOM Selling Sciences Program (AXIOM SSP)**

*Bob Nicols, AXIOM Sales Force Development*

AXIOM has over 15 years of history in this field and over that period it has become synonymous with excellence in sales performance. Bob's presentation will be based on the industry-acclaimed AXIOM Selling Sciences Program (AXIOM SSP). AXIOM SSP goes beyond "selling techniques" and illustrates how to implement a logical, repeatable, scientific process for selling. The session will include excerpts from several aspects of the process including prospecting, qualifying, presenting, negotiating, forecasting, plus complex environments and sales management. AXIOM SSP is a proven process and has been adopted by numerous Fortune 500 Companies and hundreds of small to medium-sized businesses in North America.

1:45 - 2:30 pm

### **Best Business Practices for Resellers**

*Jeanne Leckie, The Leckie Group*

Come draw on Jeanne's extensive channel marketing experience in the telecommunications industry. Learn how to strengthen the business outcomes of your customers by improving your selling process, your approach and positioning in a multi-solution, multi-vendor environment. Achieve improved business efficiencies and team productivity with simple business processes. Discover how to uncover the gaps and realign your business and your customer's business model proficiently. This session focuses on the internal and external business matters required to execute in today's competitive marketplace.

## VOICE PEERING FABRIC WORKSHOP - ROOM: 502A

**Thursday, September 18 - Free For Enterprises and Service Providers**

The Voice Peering Fabric (VPF), the preferred marketplace for carriers and enterprises to exchange voice, video and telephony services, is pleased to organize another distinctive Voice Peering Workshop at ITEXPO West 2008!

This workshop brings to ITEXPO attendees a team of experts from the telecom, hardware and software industries. Their business strategies and technical insight will be the answer to your questions and concerns relating to voice/video routing and interconnections. The workshop panel discussions present the state of direct peering, putting you on track to harness the full potential of IP communications.

8:15 am - 9:00 am

#### **State of VoIP Peering**

Speakers: TBA

9:15 am - 10:00 am

#### **Automating Voice Bilateral Interconnections**

Speakers: TBA

10:15 pm - 11:00 am

#### **Using ENUM to Optimize VoIP Call Routing**

Speakers: TBA







## **Connect Like You Never Have.**

Stop by Booth 519, and get ready to build your  
next generation of communications solutions.

**Visit Booth 519**

1.800.388.2475 • +1.905.474.1990 • [sales@sangoma.com](mailto:sales@sangoma.com) • [www.sangoma.com](http://www.sangoma.com)





## AudioCodes Benefits

- Interoperability – over 100 VoIP Applications
- Remote Branch Voice Survivability
- Advanced IP Security
- Scalability (2 to 16,000 VoIP Channels)
- ETSI Proven Voice Quality
- Legacy TDM Voice Support
- Worldwide PSTN Homologation
- OPEX Reduction



## Application Partner Ecosystem



For further information please contact: +1-408-441-1175  
or visit our website at: [www.audiocodes.com](http://www.audiocodes.com)

**Visit Our Booth #320**



# ***“All-in-one”*** BUSINESS COMMUNICATIONS PLATFORM



## ***Mediant™ 1000 MSBG*** ***Multi-Service Business Gateway***

- Merging Voice, Data, Security, Survivability and Telephony Applications into one box
- Clear demarcation point for Service Providers
- Solves management and troubleshooting conflicts
- Built-in Enterprise class Session Border Controller
- Leverages on AudioCodes vast interoperability record
- Features Stand Alone Survivability (SAS)





## Free Keynote Speeches:



**Tuesday September 16 - 3:00pm - 5:00pm 1st Level in the Petree Hall**



Jonathan Christensen  
General Manager for  
Video and Audio



Xuedong Huang,  
General Manager,  
Microsoft Communications  
Incubations



Vikram Saxena  
CTO



**Wednesday September 17 - 3:00pm - 5:00pm 1st Level in the Petree Hall**



Sharone Ben-Levi  
VP of Marketing for  
Enterprise Enhanced  
Services



Michael Tessler  
President & CEO



Bryan Martin  
Chairman &  
CEO



**Tuesday, September 16**  
11:00am  
1st Level in the Petree Hall



Lawrence Byrd  
Director, Unified  
Communications  
Architecture



**Wednesday September 17**  
11:00am  
1st Level in the Petree Hall



Louis Marascio  
Director of Engineering  
Cisco Unified Application  
Environment



**Wednesday September 17**  
11:30am  
1st Level in the Petree Hall



Fred Zimmerman  
Director, CPE VoIP Solutions  
Communications Infrastructure  
& Voice Business



### Networking Reception

Tuesday, September 16, 2008

Welcome Reception immediately  
following Keynote Sessions

5:00 PM - Main Lobby

Sponsored By:



### Networking Reception

Wednesday, September 17, 2008

Opening of the Exhibit Hall  
4:00 - 8:00 PM

Exhibit Hall

Sponsored By:



### Toyota Giveaway

Thursday, September, 2008

Details on page 72

At the conclusion of Exhibit  
Hall Hours 4:55 PM



Drawing At 4:55 PM - MUST BE PRESENT TO WIN





**ELEPHANTS: 23 MONTHS**



**HUMANS: 9 MONTHS**



**MOBILE VIDEO APPS:  
2 Months**

**Bring mobile video  
applications to life in record  
time—with NMS Communications.**

Want to create amazing mobile video applications from the ground up? Or need complete server-level solutions to power your next mobile video project? Either way, you'll be up and running before you know it with breakthrough mobile video technology from NMS Communications.

Leading telecom companies worldwide are already using NMS solutions to fuel the success of the mobile video revolution. We're a single source for the industry's most comprehensive mobile video toolkits, video servers and gateways, and transcoder systems.

To learn how we've helped developers jump-start mobile video applications, visit **booth #106** or [www.nmscommunications.com](http://www.nmscommunications.com) and shorten the time from inception to delivery.

***Win an iPod!***



Visit our booth for a  
chance to win.

**Booth #106**



(Booth Number in Parentheses)

**TMC**

# Call Center 2.0™

## Platinum Sponsors:



**VOLT DELTA**  
HOSTED SOLUTIONS



## Gold Sponsors:



### Aculab (314)

**P: +44 (0)1908 273800**

**F: +44 (0)1908 273801**

**www.aculab.com**

Aculab offers developers and service providers a wide range of hardware and software building blocks for integration into high performance communications solutions – such as fax, conferencing, contact centers and military applications.

Products include Prosody media processing platforms – available for TDM, IP and converged environments. Prosody's capabilities include rich media processing resources, VoIP protocols and voice codecs, which meet the demands for increased security and resilience supporting a number of VoIP security standards, like Secure RTP and SIPS. DSP-based media processing platforms are available in PCI, PCIe, and cPCI formats, with optional E1/T1 digital network access encompassing SIP, H.323 and SS7. Host-based media processing software is available in a flexible and cost-effective format.

Aculab also offers GroomerII – a cost-effective solution for inter-exchange and VoIP access carriers where a media and signalling gateway with intelligent call routing between SS7 and SIP-based IP networks is required. 1U and 6U rackmount configurations offer telco grade features with SNMP, RAID and N+1 redundant power supplies ensuring essential, high availability operation.

Support is available to help developers through each stage of their product's life-

cycle, including pre-sales consultancy, technical support, training and co-marketing providing the essential mix of capabilities to meet the changing needs of the converged communications market.

### ApplianX (314)

**P: +44 (0)1908 273800**

**F: +44 (0)1908 273801**

**www.applianx.com**

ApplianX is a range of single purpose, 'plug and play' appliances designed to unite, support and speed the migration to converging technologies like TDM, VoIP, Video and Mobile. Helping enterprises reduce operational costs, extend the life of existing TDM-based equipment and take advantage of new IP-based services and endpoints. Products available are the ApplianX IP Gateway and the ApplianX Gateway for Microsoft Office Communications Server (OCS) 2007.

Based on Aculab technology, both ApplianX Gateways can reduce operational costs, are simple to use and have extensive resilience features. The gateways are available with 1, 2, 4 or 8 E1/T1 trunks, supporting over 85 network and PBX inter working protocols, including DPNSS and Q.SIG. High availability is achieved through the replication of critical components, such as dual IP traffic interfaces. Finally, a web-based GUI eases configuration and management and includes a quick five step setup wizard.

The ApplianX Gateway for Microsoft OCS is a certified 'basic hybrid' gateway

developed to handle all the functions of a standard IP gateway plus the Mediation Server function of OCS in a single device. This can significantly reduce the total cost of ownership of an OCS 2007 deployment.

### Aspect Software (CC 232)

**P / F: 888-412-7728 / 978-244-7410**

**www.aspect.com**

Aspect Software, Inc. founded the contact center industry and is now the world's largest company focused solely on unified communications for the contact center. Our all-in-one, IT-ready solutions help two-thirds of the FORTUNE Global 100 and a variety of small and medium enterprises to communicate-enable their customer service, collections and sales & telemarketing business processes.

### CosmoCom (CC 231)

**P / F: 631-940-4200 / 631-940-4500**

**www.cosmoc.com**

CosmoCom's mission is to optimize communication between organizations and their customers. Its systems are improving enterprise responsiveness and the customer experience while reducing costs for companies on every continent. CosmoCom is the global leader in Contact Center Consolidation 2.0 – the all-IP contact center strategy that helps companies to be more reachable and more responsive in a cost effective way, including those with the largest and most complex enterprise requirements.



# UNIFIED COMMUNICATIONS + CONTACT CENTER



## A perfect match.

If you're looking for a place to extract value from your unified communications strategy, look to your contact center. Our unified solutions bring customers and companies together by bringing enterprise infrastructures and contact center applications together. Ole!

Learn more about unified communications for the contact center at

**[AspectComeTogether.com](http://AspectComeTogether.com)**





(Booth Number in Parentheses)



# Call Center 2.0™

## **Fonality, Inc. (402, OS 02)**

**P: 877-366-2548**

**[www.fonality.com](http://www.fonality.com)**

Fonality is a leader in business phone systems and contact center solutions for businesses of all sizes. Used by over 5,000 companies and 100,000 end users in 100 countries, Fonality's award winning IP-PBX VoIP phone systems have connected more than 225,000,000 mission critical phone calls. The PBXtra and trixbox Pro product lines are based on Fonality's patent-pending

Anywhere Management™ Hybrid-Hosted™ architecture, plus an improved version of the popular open source Asterisk code base which has been modified to add reliability, stability and enterprise-class features. PBXtra and trixbox Pro deliver the advanced capabilities of an enterprise-class phone system for 40 to 80 percent less than traditional offerings. Fonality's fully free and open source telephony platform, trixbox CE ([www.trixbox.org](http://www.trixbox.org)), is home to one of the world's largest and fastest growing communities of open source telephony users, with over 200,000 live deployments and 125,000 new downloads each month. Fonality headquarters are in Los Angeles and company investors include Azure Capital Partners and Intel Capital.

## **Interactive Intelligence (514)**

**P / F: 317-872-3000 / 317-872-3000**

**[www.inin.com](http://www.inin.com)**

Interactive Intelligence Inc. (Nasdaq: ININ) is a global provider of unified business communications solutions for contact center automation, enterprise IP telephony, and enterprise messaging. In 1997 the company released the first "all-in-one" communications software suite offering single-platform architecture with inherent multi-channel processing

minus the cost and complexity introduced by multi-point vendors. Interactive Intelligence is a leader in the adoption of voice over IP with its release in 2002 of the first converged communications application software suite to fully incorporate the session initiation protocol (SIP). Interactive Intelligence has combined this innovation with breadth of experience, today offering its more than 3,000 global customers a comprehensive solution-set comprised of premise-based and hosted offerings, including software, hardware, consulting, support, education and implementation. The company's long-term success is backed by hundreds of industry accolades, including Software Magazine's top 500 global software and services suppliers, NetworkWorld's top 200 North American networking vendors, BusinessWeek's top 50 hot growth companies, and FORTUNE Small Business Magazine's top 100 fastest growing companies. Interactive Intelligence was founded in 1994 and employs approximately 600 people. It's headquartered in Indianapolis, Indiana and has six global corporate offices with additional sales offices throughout North America, Europe, Middle East, Africa and Asia Pacific.

## **OAISYS (CC 337)**

**P / F: 888-496-9040 / 480-496-9015**

**[www.oaisys.com](http://www.oaisys.com)**

OAISYS® is a premier developer of call recording solutions for small- to medium-sized businesses and call centers. Our solutions help companies within a variety of industries — including healthcare, automotive dealerships, financial services and manufacturing — attract and retain customers by digitally capturing phone-based interactions for simple retrieval, playback and management. We offer superior integration with leading business phone sys-

tems, ensuring the right fit for a broad array of organizational needs.

## **Stealth Communications (CC 335)**

**P / F: 212-232-2020 / 212-232-2021**

**[www.stealth.net](http://www.stealth.net)**

Based in New York City, Stealth Communications specializes in peering and interconnection services for the communications industry. Stealth's "Route at Layer-2" service model promotes simplicity, speed and security by moving voice, video and data traffic between organizations in a peer-to-peer fashion.

## **Syntellect (CC 333)**

**P / F: 602-789-2803 / 602-789-2768**

**[www.syntellect.com](http://www.syntellect.com)**

At Syntellect, we help our customers create, maintain and continuously improve superior end-to-end service for their customers - personalized service that values their preferences from the way they contact a business, to the level of help desired. We measure our success by our ability to implement and continuously support solutions for our customers. With more than twenty years of pioneering leadership and thousands of solutions deployed globally, Syntellect is a premier provider of enterprise-class contact center solutions for the high-technology, financial services, help desk, utilities, government, and consumer products industries.

Syntellect, headquartered in Phoenix, Arizona, has additional offices in North America and the United Kingdom.

## **TouchStar (CC 331)**

**P: 303-338-0200 x204**

**F: 443-338-0670**

**[www.touchstar.com](http://www.touchstar.com)**

TouchStar develops and supports world class call center software with on-site and



## Benefits of CosmoCall Universe

### Complete

Satisfy all of your ACD, IVR, CTI, agent desktop, reporting, recording and outbound dialer requirements on a single, unified, virtual contact center

### Easy to Use

- **Agents** – Simple, web-based, easy to learn interface
- **Administrators** – intuitive web-based tools for administration, reporting, and call flow across your entire virtual contact center
- **IT Staff** – consolidation and multi-tenancy means one platform to install, maintain and integrate for all call centers in all locations

### Easy to Integrate

- Easily integrate one or more CRM's
- Data flows with calls for screen pop every time
- Unified agent desktop
- Integrate **once** for all locations & media

### Scalable and Reliable

- Scales up to carrier-sized systems
- Scales down for small enterprises
- Field-proven for mission-critical contact centers and telcos

### Flexible

- **Agents anywhere** – formal call center, knowledge workers, at home agents, mobile workforce, even outsourcers on the same platform
- **Communicate your way** – telephone, email, voice mail, and web chat/voice/video/collaboration
- **Deployment choices** – your premises or service provider hosted



## About CosmoCom

CosmoCom's mission is to optimize communication between organizations and their customers. Its systems are making enterprises more reachable and responsive, improving the customer experience, and reducing costs for companies on every continent. CosmoCom is the global leader in Contact Center Consolidation 2.0 the all-IP contact center strategy that improves efficiency and cost effectiveness for all companies, especially those with the largest and most complex requirements.

CosmoCom was founded in 1996 to create the next generation of contact center technology and holds several US and international patents on its multi-channel, all-IP contact center architecture. Headquartered in Melville, New York, CosmoCom has established a significant worldwide presence with offices in the US, the Netherlands, UK, Germany, France, Japan, China, Hong Kong, Brazil, India, Israel and Singapore, and distribution partners in many other countries.

## Benefit from Consolidation 2.0

Consolidation 2.0 encompasses everything: all contact center functions; all locations including onshore, offshore, and home; formal and informal agents; captive and outsourced operations; multi-channel communication; and multiple applications. CosmoCom technology enabling Consolidation 2.0 is available as a premise-based solution within the enterprise, and as a service from top global service providers hosting it.

### Consolidation 2.0

#### One Platform

- All Functions
- All Channels
- All Locations
  - Onshore & Offshore
  - Agents at Home
  - Formal & Informal
  - Captive & Outsourced
- All Applications

Service Provider Hosted or Self-Hosted

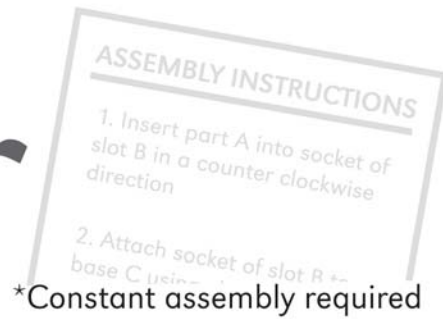
CosmoCom customers include Fortune-class, reachable enterprises throughout the world and service providers such as BT, Deutsche Telekom, Orange Business Services, Telefonica, NTT, Tata Communications, PLDT, Verizon, KPN, and many others. CosmoCom is the most-selected provider of hosted contact center platforms to top-tier telcos worldwide.





# PRE-INTEGRATED VoIP APPLICATIONS

or



[www.inin.com](http://www.inin.com)



Legacy vendors should put them right in the box, the PBX assembly instructions to add features like recording, conferencing, ACD queuing, and messaging. We simply put all the features your business needs in one IP PBX application suite to eliminate the high price and headaches of integration projects, selection of multiple vendors, and learning yet another administrative interface. That's why it's called "all-in-one" and that's why our attorneys let us say "Complete, out-of-the-box." No fine print.

**Visit us at Internet Telephony West in Los Angeles on  
September 16-18  
for a personal demo of our all-in-one IP solutions.**

***Booth # 514***



**INTERACTIVE INTELLIGENCE®**

*Deliberately Innovative*



(Booth Number in Parentheses)

# **TMC** Call Center 2.0™

hosted deployment options. TouchStar serves over 2,500 businesses on six continents from its headquarters in Denver and regional offices around the world. TouchStar's unified communications products include: call center software, traditional and VoIP telecommunications systems, predictive dialers, automated voice messaging, advanced ACD & IVR, compliance management, voice recording, and an IP PBX phone system.

## **Volt Delta Hosted Solutions (306)**

**P: 877-899-9950**

**[www.voltdelta.com](http://www.voltdelta.com)**

VoltDelta Hosted Solutions brings together VoltDelta's 30 years of experience in delivering Directory Assistance to all of the major telecom companies in North American and Europe with our state of the art contact center platform, applications and functions.

At Contact Center 2.0 VoltDelta Hosted Solutions is announcing OASIS On-Call: hosted range integrated products

and platform for automated speech and contact center services, designed to enable organizations to achieve rapid implementation, take up and usage via a per seat and or minute model without high capital investment. Some of the products in the On-Call portfolio include:

OASIS Voice Self-Service is a next generation IVR solution including CTI functionality, Speech Recognition + TTS options, and hosted voice applications built to reduce agent call time and increase caller satisfaction.

OASIS Web Station comprises solutions that enable organizations to extend agent functionality regardless of where they are located and integrate into existing software and systems.

VoltDelta is a global systems and integration company with offices in New York, California, Pennsylvania, Germany, the

Netherlands and the United Kingdom.

Visit us at booth number 306 and let us show you how VoltDelta Hosted Solutions can increase your center's effectiveness and bottom line without incurring the major capital expense.

## **Zeacom Communications Center (520)**

**P / F: 800-513-9002 / 800-513-5874**

**[www.zeacom.com](http://www.zeacom.com)**

Zeacom has successfully pioneered award-winning software applications that deliver Unified Communications solutions for organizations with up to 2500 desktops, and contact centers with between 5 and 500 agents. More than 2500 SMBs across 25 countries use Zeacom's business communications software to improve communication, collaboration and productivity. For more information about lead products such as rich and mobility-enabled presence, or SIP-based conferencing, visit [www.zeacom.com](http://www.zeacom.com)

## Platinum Media Sponsors



**InfoTech.TMCnet.com**



**WiMAXtoday.TMCnet.com**



**Cable.TMCnet.com**



**OutsourceAPAC**

**Outsource2India**



## Association Partners

**SIPFORUM**



[WWW.DIDX.ORG](http://WWW.DIDX.ORG)



# Communications Traffic Control



Ask about our Multimedia Queuing, Call Back, and Skills-Based Routing modules as part of your new Unified Contact Center Solution.

See Us At Booth #520  
Chance to WIN a 42" HD TV!



800-513-9002

usasales@zeacom.com



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## 360networks (Mtg Rm 501B)

**P / F: 206-239-4360 / 877-349-2949**

**[www.360networks.com](http://www.360networks.com)**

360networks is a leading wholesale provider of communications products and services in the western United States. Our comprehensive range of voice, private line circuits and data solutions is built on an integrated network architecture that is wholly owned and operated by the company – enabling us to provide reliable, responsive customer service direct from the source. Our solid, fiber-optic backbone is built from the ground up for consistent performance over 17,200 route miles. We currently offer a long-haul and metro presence in 48 U.S. markets, with plans to expand coverage in the future.

## 8x8, Inc. (407)

**P: 408-727-1885**

**[www.packet8.net](http://www.packet8.net)**

8x8, Inc. (Nasdaq: EGHT) offers voice and video Internet-based telephony services for business and residential customers. Marketed under the Packet8® brand name, these hosted communications solutions - Virtual Office iPBX, Complete Contact Center, Virtual Trunking and Hosted Key System - enhance the value and functionality of existing broadband Internet connections by delivering advanced features and digital quality phone service at a fraction of the cost of legacy, copper wire alternatives.

The Packet8 Virtual Office® service, currently in use by over 12,000 companies, eliminates the need for traditional business phone systems by delivering all telephony services over managed or unmanaged Internet connections. This affordable, easy-to-use alternative to traditional PBX systems or Centrex class services allows high-speed Internet users anywhere in the world to be part of a virtual PBX that includes automated attendants, conference bridges, extension-to-extension dialing and ring groups, in addition to a rich variety of other business class features normally found on dedicated PBX equipment. The Packet8 Complete Contact Center is a multimedia distribution and management platform that works with any broadband Internet service and provides enterprise class contact center functionality combined with Virtual Office hosted iPBX calling features and Packet8's cost efficient business VoIP calling plans. Packet8

Virtual Trunking services deliver a broadband agnostic IP trunking solution that allows businesses to benefit from the flexibility and cost efficiencies of VoIP (Voice over Internet Protocol), while retaining their existing phones and phone system hardware and optimizing the use of their existing connectivity to the Internet. The Packet8 Hosted Key System solution provides businesses accustomed to shared line appearance functionality with an opportunity to migrate to VoIP without altering the features and user behavior they have traditionally relied upon.

## Aastra Telecom (OS 07 & 08)

**P / F: 905-760-4200 / 905-760-4233**

**[www.aastratelecom.com](http://www.aastratelecom.com)**

Aastra, at the forefront of the Enterprise Communications market, manufactures PBX and IP PBX systems, terminals and applications addressing all market segments.

Visit us in the Open Source Pavilion to see our new MBU 400 SIP-DECT mobility solution for Small Business and our AastraLink Pro™ 160 IP Key System offering significant new feature enhancements. Aastra will also be demonstrating our recently launched AastraLink RP™, a Microsoft® Response Point™ Phone System for Small Business, a new hosted services offering and a variety of XML browser enhancements. Aastra continues to offer powerful, innovative products and services for Enterprise and SMB markets.

## ABP Technology (618)

**P / F: 972-831-1600 / 972-831-1416**

**[www.abptech.com](http://www.abptech.com)**

ABP Technology is value-added distributor of IP Telephony and IP Video solutions and services. Based in Dallas, TX. ABP supports hundreds of resellers and Service Providers throughout the Americas and beyond. The ABP portfolio includes a broad variety of IP communication solutions and peripherals like IP Phones, IP paging products, Analog and Digital gateways. Mobility options include GSM gateways, Wi-Fi, DECT and Dual mode phones. Our Video technology includes SIP based Surveillance Cameras, video IP Phones, special software for facial, license plate and bar code recognition, video analytics and pre-sales support. An entire line of peripheral accessories are available

from ABP including NAS, UPS, WiFi antennas, PoE switches and extenders, Internet redundancy products and PSTN failover. ABP provides engineering consultation services, technical support, IP training along with provisioning, configuration and fulfillment services. ABP offers a suite of IP Carrier trunking and connectivity to complement our resellers' hardware solutions. ABP's clients include channel partners, specialized VARs, Service Providers, government entities and educational institutions.

## Aculab (314)

**P: +44 (0)1908 273800**

**F: +44 (0)1908 273801**

**[www.aculab.com](http://www.aculab.com)**

Aculab offers developers and service providers a wide range of hardware and software building blocks for integration into high performance communications solutions – such as fax, conferencing, contact centers and military applications.

Products include Prosody media processing platforms – available for TDM, IP and converged environments. Prosody's capabilities include rich media processing resources, VoIP protocols and voice codecs, which meet the demands for increased security and resilience supporting a number of VoIP security standards, like Secure RTP and SIPS. DSP-based media processing platforms are available in PCI, PCIe, and cPCI formats, with optional E1/T1 digital network access encompassing SIP, H.323 and SS7. Host-based media processing software is available in a flexible and cost-effective format.

Aculab also offers GroomerII – a cost-effective solution for inter-exchange and VoIP access carriers where a media and signalling gateway with intelligent call routing between SS7 and SIP-based IP networks is required. 1U and 6U rack-mount configurations offer telco grade features with SNMP, RAID and N+1 redundant power supplies ensuring essential, high availability operation.

Support is available to help developers through each stage of their product's life-cycle, including pre-sales consultancy, technical support, training and co-marketing providing the essential mix of capabilities to meet the changing needs of the converged communications market.





## HOW RELIABLE IS YOUR VOIP E911 SERVICE?

With DASH911, reliability is a key component. We take care of the details so your company can focus on its core business.

- Best enhanced nationwide coverage of any E911 provider
- Easy integration. Simple provisioning
- Flexible SIP or PSTN interconnect
- Unmatched 24x7 support by E911 experts
- Proven partner. More service providers choose DASH911 than any other E911 provider
- Affordable service. Lowest "Total Cost of Ownership" for FCC compliant E911
- Patent-pending technology
- Complete CDR, reporting, and management tools

**Contact a Dash Carrier Services representative TODAY!**

**303-228-8800**

**[sales@dashcs.com](mailto:sales@dashcs.com)**



**DASH**  
*carrier services*  
[www.dashcs.com](http://www.dashcs.com)

**Visit us at ITEXPO**  
**September 16-19, 2008**  
**Booth #622**



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## **ADTRAN, Inc. (612)**

**P / F: 256-963-8000 / 256-963-8699**

**[www.adtran.com](http://www.adtran.com)**

ADTRAN offers a suite of IP business solutions for converged IP networking. These include a variety of business trunking, hosted VoIP and premises-based VoIP solutions including multiservice gateways, access routers, managed Layer 2/3, PoE and Gigabit Ethernet switches; IP PBX systems; IP phones; 802.11 a/b/g Wireless Access Points; and security/VPN appliances. These products are ideal for bundled services or business networks and address the needs for branch office connectivity, Internet access, VoIP migration, bandwidth expansion and network security.

## **Airespring (525)**

**P / F: 818-786-8990 / 818-786-9225**

**[www.airespring.com/sip](http://www.airespring.com/sip)**

Headquartered in Los Angeles, award winning AireSpring is one of the fastest growing telecommunications carriers in the USA, with over 3 billion call records processed every year. AireSpring is a privately held company offering a broad range of innovative services to corporate and residential customers nationwide. Information about AireSpring's SIP products can be found at [www.airespring.com/sip](http://www.airespring.com/sip).

## **AItech (537)**

**P / F: 952-829-5511 / 952-829-5601**

**[www.goait.com](http://www.goait.com)**

AItech is a leading provider of wholesale & retail hosted PBX, IP-PBX and managed IT solutions. AItech's suite of managed network services offer a highly effective solution to address residential and business end-user needs, including the ability to use Outlook® and Internet Explorer® to dial/answer calls and change telephone settings

## **Altai Technologies (618)**

**P / F: + 852 2116 8087 / + 852 2607 4021**

**[www.altaittechnologies.com](http://www.altaittechnologies.com)**

Altai Technologies is the leading wireless broadband solution provider. Its award-winning product, A8 WiFi Base Station is deployed in large area/outdoor environment to build a secure and robust WiFi network with unprecedented range. Utilizing innovative wireless technologies, A8 effectively provide much larger coverage than other outdoor WiFi base stations.

## **APEX Voice Communications (531)**

**P / F: 818-379-8400 / 818-379-8410**

**[www.apexvoice.com](http://www.apexvoice.com)**

APEX is a supplier of IMS-compliant, multi-service SIP Application Servers for voice, video and presence-aware enhanced services to over 250 wireless and wireline network operators and service providers across 95 countries. Applications are quickly and efficiently developed using either the browser-based service creation environment or Python, and managed using the integrated OAM&P module.

## **ApplianX (314)**

**P: +44 (0)1908 273800**

**F: +44 (0)1908 273801**

**[www.applianx.com](http://www.applianx.com)**

ApplianX is a range of single purpose, 'plug and play' appliances designed to unite, support and speed the migration to converging technologies like TDM, VoIP, Video and Mobile. Helping enterprises reduce operational costs, extend the life of existing TDM-based equipment and take advantage of new IP-based services and endpoints. Products available are the ApplianX IP Gateway and the ApplianX Gateway for Microsoft Office Communications Server (OCS) 2007.

Based on Aculab technology, both ApplianX Gateways can reduce operational costs, are simple to use and have extensive resilience features. The gateways are available with 1, 2, 4 or 8 E1/T1 trunks, supporting over 85 network and PBX inter working protocols, including DPNSS and Q.SIG. High availability is achieved through the replication of critical components, such as dual IP traffic interfaces. Finally, a web-based GUI eases configuration and management and includes a quick five step setup wizard.

The ApplianX Gateway for Microsoft OCS is a certified 'basic hybrid' gateway developed to handle all the functions of a standard IP gateway plus the Mediation Server function of OCS in a single device. This can significantly reduce the total cost of ownership of an OCS 2007 deployment.

## **Asteria Solutions Group (OS 01)**

**P / F: 256-705-0277 / 256-705-0280**

**[www.asteriasgi.com](http://www.asteriasgi.com)**

Asteria Solutions Group develops commercial software for call centers, automat-

ed dialing, and business PBXs, and offers product maintenance and support.

Asteria also provides services to develop and support custom applications built on Asterisk®. Expertise with open source software coupled with extensive experience in telephony enables lower cost solutions for clients. Custom solutions deployed include messaging and notification systems, custom IVRs, prepaid calling applications, provisioning and activation systems, and custom conferencing platforms.

## **AudioCodes (320)**

**P / F: 408-441-1175 / 408-451-9520**

**[www.audiocodes.com](http://www.audiocodes.com)**

AudioCodes (NASDAQ: AUDC) provides innovative, reliable and cost-effective Voice over Packet technology and Voice Network products to OEMs, network equipment providers and system integrators. AudioCodes offers its customers and partners a diverse range of flexible, comprehensive media gateway and media processing technologies, based on VolPerfect? ? AudioCodes? underlying, best-of-breed, core media gateway architecture. The company is a market leader in voice compression technology and is a key originator of the ITU G.723.1 standard for the emerging Voice over IP market. AudioCodes? international headquarters and R&D facilities are located in Israel, with U.S. headquarters in San Jose, California.

## **Avaya - DevConnect Program (ComDev 114)**

**P / F: 908-953-6000 / 732-852-1397**

**[www.avaya.com/devconnect](http://www.avaya.com/devconnect)**

The Avaya DevConnect Program provides a wide range of free developer resources, including Avaya APIs and SDKs, developer tools, technical support and education to those interested in designing Avaya-compatible solutions. Enhanced Membership options offer increased technical support, compliance testing, and co-marketing benefits. To register for membership, please visit [www.avaya.com/devconnect](http://www.avaya.com/devconnect).

## **Basis Audionet (313)**

**P / F: 215-674-8600 / 215-674-5714**

**[www.basis-audionet.com](http://www.basis-audionet.com)**

Basis Audionet – A global Telecommunications Leader

Providing IP services to resellers, carriers, ISPs and MSOs. With a proven track record of success in IP, mobile, and land-



**Help you create and dominate future!** With continuous innovation and finest design, Synway leads you to the right track, and empowers you to deliver substantial values. Decades-of-years design expertise and complete product ranges deliver eternal values for your Today and future's growth. We enable you to exceed market expectation in IP and PSTN-based services, including MMS, WAP, SMS, CRBT, Prepaid card, Conferencing, Media Gateway and more.

**Synway's media processing and signaling technologies** feature high interoperability, scalability, customizability and reliability, and help you deliver innovative convergence solutions with matchless price-performance ratio. You will experience endless excitements, and overcome overwhelming challenges.

**World-class service and technical support** will be at your fingertip. Striving to establish eternal eco-relationships with application developers and service providers, Synway will commit to remove your complexities and creating unique experience at present as well as for future.

For more details, please call us at (86) 571 8886 0561 or visit us at <http://www.synway.net>

**See us at booth 117 at ITEXPO WEST 2008**







- *Virtual Agent*
- *"Agent at Home" Solutions*
- *Speech Automated IVR*
- *Advanced ACD*
- *Outbound Alerts*
- *Transport Solutions*
- *Conference Calling Solutions*
- *Call Recording*
- *Agent Monitoring and Observation*
- *Reporting*
- *Computer Telephony Integration*
- *Hosted On-Demand Model*
- *Complete, Customized Solutions*

1 Sentry Parkway, Suite 6000  
Blue Bell, PA 19422  
610.276.4300

[www.VoltDelta.com/VoltDelta-Hosted-Solutions](http://www.VoltDelta.com/VoltDelta-Hosted-Solutions)  
[info@VoltDelta.com](mailto:info@VoltDelta.com)

**VoltDelta - over 9 billion inbound calls served already...  
...and if you have ever called 411 your call was too.**

VoltDelta is the market share leader in Directory Assistance – and now we are making the technologies that power 411 offerings available to all contact centers.

*Introducing VoltDelta Hosted Contact Center Solutions.*

### **Virtual Agent:**

Hosted browser session provides access to any contact center application  
Soft phone to perform all call handling, dialing, conferencing and call transfer  
Supports SIP Phones, Dedicated telephone numbers / PBX extensions  
Call Recording and Agent Monitoring and Observation

### **Hosted, Speech Automated IVR:**

Complete speech recognition solutions; many pre-developed; customizable applications  
Integration to back end systems & databases  
Text to speech  
Rich development feature sets to easily deploy speech solutions

### **Advanced Switching:**

Fully hosted ACD  
Intelligent routing based on desired business rules  
Integrates with a variety of transport options  
Route calls anywhere in the world

### **Managed Transport:**

Private, managed, VoIP Network  
Global footprint  
Seamless linking of multiple contact centers  
SIP functionality

### **The VoltDelta Hosted Solutions Difference**

VoltDelta works with our customers to understand their exact current and future needs and to tailor and deliver the best solution for them. By leveraging our 30 years of implementing the largest and most sophisticated contact centers for the major telecommunications companies and our hosted, on-demand model, centers of all types can take advantage of the VoltDelta Hosted Solutions advantage.

Customized solutions  
Per minute used pricing  
Minimal up-front costs  
Quick implementation times





## Fully Hosted

## On-Demand

## IP Contact Center

OASIS On-Call, the foundation of VoltDelta's Hosted Contact Center Solutions, is a completely hosted range of integrated products and platform for automated speech and contact center services, designed to enable organizations to achieve rapid implementation, take up and usage without high capital investment.

OASIS Voice Self-Service is a next generation IVR solution including CTI functionality, Speech Recognition + TTS options, and hosted voice applications built to reduce agent call time and increase caller satisfaction.

OASIS Web Station comprises solutions that enable organizations to extend agent functionality regardless of where they are located and integrate into existing software and systems.



Call Routing  
Speech Enabled  
Conferencing  
Alerting  
Response Capture  
Call Back  
Automated Survey  
Verification  
Payment



Agent Anywhere  
Functionality  
Agent Browser  
Softphone  
CRM Integration  
RTA & Historical  
Reporting  
Agent Management  
Multi-Channel

Universal Routing  
Virtual Queue  
Personalization  
SIP Enabled  
Media Gateways



Voice Automation  
CTI Reporting  
Quality Management  
Voice Recording  
& Screening



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

line communications integration, our company provides solid system implementation expertise. Voice chat, FMC, signaling, switching, Media Gateway and application server solutions are just a few of the many areas in which Basis Audionet specializes.

Offering a broad palette of customer relations management services, while combining core competencies in IT and value-added IP communications solutions development, Basis Audionet is dedicated to excellence and committed to quality service, as your technology partner of choice.

**BillCall - A Panamax Infotech Division (142)**  
**P / F: 982-502-9183 / 793-011-7675**  
[www.billcall.net](http://www.billcall.net)

BillCall is a hosted as well as on premise billing service for wholesale telecom carriers. It provides a full-featured billing system via your secure Web browser, while eliminating the burden of maintaining costly on-site billing hardware, software, and staff.

A Solution that has the following:

- Rate Management
- Route Management
- Reports
- Alerts
- Invoicing & Accounting
- Client Management
- Resource Management

In all it is a complete solution for your soft-switch, Gateway & Session Border Controller

**Business Solutions (253)**  
**P / F: 814-897-9000 / 814-868-2576**  
[www.businesssolutionsmag.com](http://www.businesssolutionsmag.com)

Business Solutions magazine inspires executives of IT channel companies (VARs, integrators, and solution providers) to grow their businesses by teaching them how to sell new IT products and services, penetrate thriving vertical markets, and integrate complementary technologies. Business Solutions delivers its insight through channel success stories and advice from industry analysts, vendors, and our expert technology editors.

**CamrivoX (618)**  
**P: +44 1223 750050**  
**F: +44 8700 689910**  
[www.camrivoX.com](http://www.camrivoX.com)

CamrivoX is an award winning technology leader delivering products for unifying telephony with On Demand CRM applications. Focused on small and mid-sized

businesses, the company's Flexor™ family of CTI software provides a simple way to integrate telephony with applications at the desktop bypassing costly system and equipment upgrades.

**Cbeyond (438)**  
**P / F: 678-424-2400 / 678-424-2500**  
[www.cbeyond.net](http://www.cbeyond.net)

Cbeyond, Inc. (NASDAQ: CBEY) is a leading IP-based managed services provider that delivers integrated packages of communications and IT services to more than 36,000 small businesses throughout the United States. Cbeyond offers more than 30 productivity-enhancing applications including local and long-distance voice, broadband Internet, mobile, BlackBerry(R), broadband laptop access, voicemail, email, web hosting, fax-to-email, data backup, file-sharing and virtual private networking. Cbeyond manages these services over a private, 100-percent Voice over Internet Protocol (VoIP) facilities-based network.

**ChannelVision Magazine (Beka Publishing) (650)**  
**P / F: 480-503-0770 / 480-503-0990**  
[www.channelvisionmagazine.com](http://www.channelvisionmagazine.com)

ChannelVision is a bi-monthly magazine read by channel partners selling voice, IP, conferencing, Ethernet, data and access services to business customers, primarily in the small/medium business space. Over 77% of ChannelVision's subscribers are telecom agents/dealers/distributors and VARs. The rest 23% are Carriers and NSPs. Visit [www.channelvisionmagazine.com](http://www.channelvisionmagazine.com) for your FREE subscription.

**Cisco (ComDev 120)**  
**P: 408-526-4000**  
[www.cisco.com](http://www.cisco.com)

Cisco is the worldwide leader in networking that transforms how people connect, communicate, and collaborate. Since 1984, Cisco has led in the innovation of IP-based networking technologies, including routing, switching, security, TelePresence, unified communications, video, and wireless. The company's responsible business practices help ensure accountability, business sustainability, and environmentally conscious operations and products.

**ClearOne (618)**  
**P / F: 801-975-7200 / 801-977-0087**  
[www.clearone.com](http://www.clearone.com)

ClearOne is a leading global provider of premium audio conferencing systems and

other related products for audio, video and web conferencing applications. The reliability, flexibility and performance of our comprehensive solutions save organizations time and money by creating natural environments for effective and efficient group communication

**Commetrex Corporation (ComDev 127)**  
**P / F: 770-449-7775 / 77-0242-7353**  
[www.commetrex.com](http://www.commetrex.com)

Commetrex develops and markets enabling technologies for the telecom OEM. BladeWare fax solutions, including Fax Media Server, Fax2Email & Email2Fax BladeWare voice solutions, including BladeWareVXML Interpreter, BladeWareMRCP, and BladeWareStudio, application creation tool. Media Processing, such as T.38 Relay, TerminatingT38, and V.34. Solutions that are built with proven, award-winning products that push the edge of technology to deliver a truly best-of-class offering for the enterprise as well as carrier OEMs, system developers, and VARs. At Commetrex, it's what we do.

**Critical Links (432)**  
**P / F: 973-276-9006 / 973-276-9007**  
[www.critical-links.com](http://www.critical-links.com)

Critical Links' edgeBOX unified communications solution provide Fortune 1000 level companies with voice, data and IT features, in an all-in one-device, at SMB prices. Whether your staff is at the main office, branch location or tele-commutes, edgeBOX's powerful interface is easy to manage while providing secure communications anywhere in the world. Supporting up to 300 users per edgeBOX, this scalable solution can be expanded, without limits, as your business grows.

**CyberData Corporation (131)**  
**P: 831-373-2601**  
[www.cyberdata.net](http://www.cyberdata.net)

CybererData manufactures peripheral devices for VoIP phone systems that facilitate:

- Application specific endpoints for Unified Communications
- IP equivalents for existing analog devices
- Legacy migration to VoIP
- OEM design and manufacturing of VoIP related products.

Our IP Speakers, Gateways, Intercoms, and UC Authentication Devices are cost-effective and easy to manage solutions for VoIP phone systems.





BEST TECHNOLOGY > BRIGHT IDEAS > CLEAR FUTURE



Visit us at  
**BOOTH 619**

to find out why TMC  
has declared Tmedia™  
**PRODUCT OF THE YEAR**  
and "BEST OF SHOW" for  
Service Provider Solutions.

## The only serious choice for Telecom Solution Providers

needing reliable, highly scalable and efficient telecommunications solutions.

**Tmedia™**



Many of the largest and most successful telecom solution providers in the world are turning to **TelcoBridges**, and our award-winning **Tmedia™** family of hardware and media gateway solutions.

Witness firsthand how the **Tmedia™** family is setting new benchmarks for performance, scalability, reliability, and efficient power consumption.

Anyone seeking to learn more about the latest advancements in developing high-availability systems should attend the TelcoBridges-led session in the  
**IT EXPO Communications Developer Conference**  
Tuesday, September 16th at 1:45 PM (Room CD2-05).

Stop by Booth 619 to obtain passes for this informative conference session.

### MAKE THE MOST OUT OF YOUR EXPERIENCE AT IT EXPO

by seeing for yourself why telecom solution providers - in increasing numbers - are turning to **TelcoBridges** to meet their needs for scalable, available, reliable, and efficient telecommunications solutions.

[www.telcobridges.com](http://www.telcobridges.com)

Photo © Stephen Coburn, Mwproductions | Dreamstime.com



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## Cylogistics (326)

**P / F: 650-694-4949 / 650-694-4953**

**[www.cylogistics.com](http://www.cylogistics.com)**

Cylogistics is a specialty distributor featuring VoIP products from Ring Carrier, Lan Power Systems & TrueBill.

Ring Carrier is a complete integrated hosted solution for LECs, CLECs, & ITSPs. It allows you to scale from 1 user to millions. It supports a Provider, Enterprise and end-User interface.

LAN Power Systems invents Power over Ethernet Products that are feature-built for modern Networks. LAN Power Products exceed today's PoE specifications and your expectations.

TrueBill is an IP billing system that includes VoIP billing. It is a very comprehensive package with a very easy to use interface. It supports PrePaid, Post Paid, Wholesale, Agents, Taxes and much more.

## Dash Carrier Services (622)

**P / F: 303-228-8800 / 303-228-8801**

**[www.dashcs.com](http://www.dashcs.com)**

Dash features a complete line of VoIP solutions that enable carriers to quickly and easily deploy new services at minimal cost. With Dash supporting your efforts, you can focus on growing your core business while leveraging our expertise to deliver the VoIP service your customers need. By partnering with Dash, you can profit from our:

**Dash911** - e911 for VoIP - The premier e911 solution for VoIP enables VoIP service providers to quickly and affordably deliver reliable FCC compliant 911 service to their customers.

**DashConnect** - Origination and Termination - Direct Inbound Numbers (DID) in over 300 US and International major markets including 6000+ US rate-centers. Nationwide termination and International call completion to over 200 countries worldwide.

**DashInfo** - 411 Directory Assistance and CNAM Delivery - Accurate, live operator answered service that helps services providers deliver the complete solution. Deliver Calling Name caller-id for any inbound call.

Dash understands that service providers need more than just good pricing. We

deliver reliable voice services backed by exceptional support. Experience the Dash difference and understand why Dash is the clear choice in carrier services.

## Dialexia Communications Inc. (433)

**P / F: 514-693-8500 / 514-693-5352**

**[www.dialexia.com](http://www.dialexia.com)**

Dialexia Communications is a next generation communication software and services company established in the Techno-Park area of Montreal, Canada.

Dialexia develops and markets a complete suite of IP Telephony software and applications. It also provides Advanced Telephony Solutions using Internet Protocols to enterprises, service providers as well as telephony resellers.

As one of the pioneers in the Voice over IP (VoIP) industry, Dialexia has implemented the latest technologies to facilitate the convergence of Voice and Data networks.

In the short term, potential SME's will be able to make a significant transition towards a telecommunication system that will provide enhanced productivity by simplifying the complexity of their numerous needs: one system will be enough to manage telephony, Internet, e-mail and long distance thanks to Dialexia's low cost VoIP solutions.

## Dialogic (SPONSOR)

**P / F: 800-755-4444 / 973-967-6262**

**[www.dialogic.com](http://www.dialogic.com)**

As a company committed to growth, Dialogic has become a leading provider of world-class technologies based on open standards such as host media processing and other multimedia products that enable its customers and partners to deliver innovative mobile, video, IP, and TDM solutions for Network Service Providers and Enterprise Communication Networks. Decades of experience and continued investment uniquely qualify Dialogic to create leading-edge products, and its partners rely on these flexible components to rapidly deploy value-added solutions around the world.

## DIDX & Super Technologies (154)

**P / F: 850-207-2598 / 206-202-0743**

**[www.didx.net](http://www.didx.net)**

Super Technologies, Inc. since 1999 focuses on enabling wired and wireless carriers as well as Web 2.0s to go global with services such as DIDX, CFMS,

and Hosted VoIP. The DIDXchange, trusted by 10,043 ITSPs, offers an API for immediate buying and selling of direct inward dialing phone numbers from 55 countries with LNP in USA, CA, France and UK as well as QoS of vendor rating and first 36 hours free trial of DID. New revenues and more satisfied customers for all service providers is Super Technologies' goal.

## DIGITALK Ltd (207)

**P / F: +44 1908 425000 / +44 1908 425001**

**[www.digitalk.com](http://www.digitalk.com)**

The DIGITALK Multiservice Platform offers a highly scalable, high performance, high availability service platform, operating from a small footprint solution to a large scale multi-node deployment. DIGITALK Broadband Telephony service-ready solution enables deployment in switched and Next Generation Network (NGN) deployment environments and providing full OSS and BSS support.

## Digium (OS 06)

**P / F: 256-428-6000 / 256-864-0464**

**[www.digium.com](http://www.digium.com)**

Switchvox® is everything that you don't expect from a PBX. It's truly affordable, easy to set up, simple to configure, and a breeze to maintain.

It has features that let your business run more effectively and with fewer hassles. And it does all of this for a fraction of the cost of the PBX dinosaurs of the past.

Switchvox is so much more than just an office phone system. It's a revolution in business communications, putting you in control of your most important asset in business, your voice.

With this incredible leap in technology comes astounding cost-savings for your business, integration capability that you never thought possible, and the flexibility to meet the needs of whatever industry that you're in.

## emFAST Inc. (147)

**P / F: 866-436-3278 / 416-208-6770**

**[www.emFAST.com](http://www.emFAST.com)**

emFAST Inc.'s FACSys® Fax Messaging Gateway solution has been a driving force in the network fax server industry since 1989. Its current platform leverages the Fax-over Internet Protocol (FoIP), allowing organizations to consolidate their infra-



“The secret of  
**Success**  
is to know something  
nobody else knows.”

—Aristotle Onassis

**CELEBRATING 100 SERVICE PROVIDER DEPLOYMENTS SINCE  
OUR FIRST COMMERCIAL SHIPMENT IN APRIL 2004!**



**Innovative Solutions for VoIP Service Providers**



Get to know Sansay and the amazing, award-winning VSX and SPX session controllers for VOIP-based service providers: visit us at [www.sansay.com](http://www.sansay.com) and meet us live at Internet Telephony Expo.






## Day One: Tuesday – September 16, 2008

	<b>IPTV Workshop for Carriers</b>	<b>TMC University: Microsoft OCS</b>	<b>TMC University: FMC/Mobility</b>	<b>TMC University: Open Source</b>	<b>Call Center 2.0 at ITEXPO</b>	<b>TMC University: SIP in the Contact Center</b>	<b>Ingate's SIP Trunking Workshop</b>
	<i>Conference fee required</i>	<i>Conference fee required</i>	<i>Conference fee required</i>	<i>Conference fee required</i>	<i>Conference fee required</i>	<i>Conference fee required</i>	<i>Free for All Attendees</i>
<b>ROOM</b>	<b>510</b>	<b>512</b>	<b>513</b>	<b>514</b>	<b>511A</b>	<b>511BC</b>	<b>507</b>
8:30	Continental Breakfast - Paid Attendees Only						<b>inGate</b>
9:00 - 9:45	How to Implement IPTV Networks	Intro	Overcoming FMC Challenges	Advantages of Open Source VoIP	Adding Intelligence to the Next-Gen Contact Center	Enabling Home-Based Agents	<b>SIP Trunking Professional Development Day</b> <b>9:00-10:30am:</b> Introduction to SIP Trunking <b>11:00am-12:30pm:</b> Enterprise Infrastructure
10:00 - 10:45	Overcoming IPTV Challenges	OCS-PBX Interoperability	Addressing Femtocell Integration Challenges	Opportunities in Deploying Open Source Applications	Contact Centers: Results from the Labs	Adoption of IP in the Next-Generation Contact Center	
11:00 - 11:45	Measuring Quality of Experience for IPTV Deployments	OCS Servers	Messaging and Mobile UC	Understanding Open Source Standards	Trends in IP Contact Center Deployment	Technology Considerations for Contact Ctr Evolution	
11:45	Conference Luncheon - Paid Attendees Only						
12:45 - 1:30	What Will it Take to Deploy IPTV?	Call Flow Scenarios	E-911 Considerations	Using Open Source to Roll Out Next-Gen Applications	Contact Center Security Strategies	Make Your Contact Center Smarter: Best Practices	<b>1:30pm-3:00pm:</b> Service Provider Perspective <i>Professional Development Certificate awarded to those who attend all three Tuesday seminars.</i>
1:45 - 2:30	The Progression of Streaming Video	Configuration & Planning	Cost Containment as a Driver for FMC Deployment	Recognizing the Opportunities in the SMB Market	Embracing Unified Communications in the Contact Center	Welcome to the Contact Center of the Future	
2:45 - 3:30	Providing differentiated services with a converged video experience: IPTV, Mobile TV, and Internet TV	TMC University Exam	TMC University Exam	TMC University Exam	Planning Ahead for Optimal Contact Center Deployment	TMC University Exam	
3:30	<b>FREE Keynote Session: Jonathan Christensen, Skype - Xuedong Huang, Microsoft - Vikram Saksena, Sonus Networks</b>						
5:00	<b>FREE Networking Reception</b>						

## Day Two: Wednesday – September 17, 2008

	Service Provider Solutions	Unified Communications	Hosted Communications	TMC University: IP Network Security	Call Center 2.0 at ITEXPO	TMC University: SaaS	Ingate's SIP Trunking Workshop
	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Free for All Attendees
ROOM	510	512	513	514	511A	511BC	507
8:30	Continental Breakfast - Paid Attendees Only						inGate
9:00 - 9:45	Quality of Service Considerations	Deploying UC	Welcome to Hosted VoIP	VoIP Security Myths & Realities	Position Your Contact Center in a Web 2.0 World	Software-as-a-Service: The Basics	<b>Mechanics of SIP Trunking</b> <b>8:30-9:45am:</b> Case Study: The Benefits of Bundled SIP Trunk Solutions <b>10:15am-11:30am:</b> SIP Trunking & Security in an Enterprise Network <b>1:15-2:30pm</b> Case Study: American Idol Gets In Tune with SIP Trunking <b>3:00-6:00pm</b> SIP Forum SIPconnect Compliance Workshop
10:00 - 10:45	The Continuing Evolution: From TDM to SIP	Network Management & QoS	Deploying Hosted VoIP	Best Practices in VoIP Security	Understanding Your Customers	Using Hosted Speech Solutions in the Call Center	
11:00 - 11:45	Applications as a Competitive Differentiator	Exploring Mobility in Unified Communications	Delivering on the Promise of Hosted VoIP	Security Considerations for the Enterprise	Benefits of VoIP Enabled Recording	CRM Software as a Service	
11:45	Conference Luncheon - Paid Attendees Only						
12:45 - 1:30	Service Creation Considerations	UC for the SMB	Open Source Options for Hosted Voice	Deploying Secure Wireless VoIP	IP Contact Center Shootout	SaaS and the Home Agent Model	
1:45 - 2:30	Imagining Tomorrow's Wireless Landscape	UC Round Table: The Future of the Industry	Future of Hosted Telephony	TMC University Exam		TMC University Exam	
2:30	FREE Keynote Session: Sharone Ben-Levi, AudioCodes - Michael Tessler, Broadsoft; - Bryan Martin, 8x8, Inc.						
4:00 - 8:00	Exhibit Hall Grand Opening Reception - Sponsored by Aculab						



# Exhibitor Descriptions

Day Three: Thursday – September 18, 2008

	Service Provider Solutions	TMC University: Unified Communications	VoIP for SMB	TMC University: SIP Hosted by Dialogic	Call Center 2.0 at ITEXPO	TMC University: Next-Gen Call Center Mgmt	Ingate's SIP Trunking Workshop
	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Conference fee required	Free for All Attendees
ROOM	510	512	513	514	511A	511BC	507
7:30	Continental Breakfast - Paid Attendees Only						inGate
8:15 - 9:00	Not All Collocation is Created Equal	Introduction to Unified Communications	Reality Check: Southern California End Users Speak Out	Introduction: SIP as the Foundation	Innovation in the Skype-Enabled Call Center	Contact Center Benchmarking Study Results	8:30-10:30am  SIP Trunk "Basic Training" with Avaya and Ing
9:15 - 10:00	Customer Retention Through Analytics	Defining the Business Case for UC		Signaling: SIP, SS7 & Integrating Your Network...	Using Natural Language to Improve the Customer Self Service Experience	Improving Customer Experience One Transaction at a Time	
10:15 - 11:00	Keeping Mobile Carriers Competitive	Unified Communications in the Call Center	SMB Networking Alternatives	Media: Voice & Video in Your SIP Environment	Service and Support: Strengthening the Bottom Line	The Importance of First Call Resolution	
11:00	Exhibit Hall Open						
12:00	Conference Luncheon - Paid Attendees Only						
12:45 - 1:30	The Convergence of Communications & Entertainment	Collaboration & Conferencing	Making the Business Case for VoIP in the SMB	Transitioning IMS via Your SIP-based Network	Extreme Customer Satisfaction: Managing Loyalty in a Commoditized Market	Leveraging Speech Analytics for Customer Satisfaction	
1:45 - 2:30	Accelerating IMS Deployment	Top 10 Ways to Get the Most From Microsoft UC	SMB VoIP Options	SIP Trunking	Contact Center Mobility	Workforce Optimization Strategies	
2:45 - 3:30	Service Provider State of the Industry	TMC University Exam	The Role of Applications	Case Study / TMC University Exam	Customer Experience 2.0: Competitive Differentiation	TMC University Exam	

## Free Keynote Speeches:



**Jonathan Christensen**  
General Manager for Video and Audio



**Michael Tessler**  
President & CEO



**Bryan Martin**  
Chairman & CEO



**Xuedong Huang,**  
General Manager,  
Microsoft Communications Incubations



**Sharone Ben-Levi**  
VP of Marketing for Enterprise Enhanced Services



**Vikram Saksena**  
CTO



### Platinum Sponsors:



### Gold Sponsors:





(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

structure and virtualize their IT environment. Find out more by visiting booth #147.

## **Epygi Technologies Ltd. (434)**

**P / F: 972-692-1166 / 972-692-1166**

**[www.epygi.com](http://www.epygi.com)**

Reliable, secure, and easy to install and use, the Quadro range of feature-packed IP-PBXs, high capacity VoIP gateways and cost-saving conference servers offer users outstanding benefits and an unparalleled range of features at very economic prices. The state-of-the-art equipment gives superior sound quality and a quick return on investment because of its low maintenance cost. Quadros are the ideal way for organizations to take advantage of the business-enhancing and money-saving facilities of VoIP telephony.

Epygi Technologies Ltd. is a privately held US company. It was founded in 2000 and has its global headquarters in Plano, Texas. It designs and manufactures its market-leading products using its own IP PBX call manager software based on standard SIP technology. Through its network of respected distributors and resellers, this global company provides converged telecom solutions to organizations wherever they are.

## **Excendia (442)**

**P / F: 514-765-8480 / 514-765-8480**

**[www.excendia.com](http://www.excendia.com)**

Excendia develops mobility solutions for enterprises and telecom service providers. Its speech-enabled Virtual Personal Communications Assistant for mobile people provides voice controlled Outlook-by-Phone, Contacts-by-Phone, Calendar-by-Phone, Smart Phone Number and Smart Fax services using ASR, TTS, SIP-based IP telephony, UC, Intelligent Call Routing and other advanced technologies previously available only to large corporations.

## **FaxBack, Inc. (533)**

**P / F: 503-597-5355 / 503-597-5301**

**[www.faxback.com](http://www.faxback.com)**

FaxBack - NET SatisFAXtion VoIP Fax Server Solutions With over twenty years of fax leadership and expertise, FaxBack's NET SatisFAXtion VoIP fax solutions are ideal for any size company or enterprise including high density carrier applications. NET SatisFAXtion enables companies to extend high availability, full feature fax into VoIP, uni-

fied messaging and desktop environments. FaxBack's VoIP fax solutions are fault tolerant, flexible and easily integrate with leading document management solutions.

## **FaxCore, Inc. (D1)**

**P / F: 729-870-2900 / 720-870-4141**

**[www.faxcore.com](http://www.faxcore.com)**

FaxCore is the only True 21st Century Fax Server product line. The product scales from small SMBs to larger enterprises. Additional, its hosted architecture makes it ideal for Software-as-a-Service (SaaS) providers plus VoIP and other hosted services. FaxCore led the way to integration with Sharp's OSA, Fuji Xerox' Apeos plus imaging products from HP, Xerox and others. These integrations facilitate workflow and enhance compliance with Sarbanes-Oxley, HIPAA and other regulatory policies. FaxCore exclusively supports fax products from Dialogic/Brooktrout including their FoIP SR140 product line. In fact, FaxCore installed the first commercial deployment of SR140 anywhere and the first SR140 deployment with Cisco's Call Manager. Stop by Booth D1 and learn more about 21st Century solutions.

## **FaxSIPit Services Inc. (535)**

**P / F: 866-374-6495 / 866-533-6734**

**[www.FaxSIPit.com](http://www.FaxSIPit.com)**

A VoIP network is not a Fax-over-IP (FoIP) network. VoIP networks can transmit faxes to some numbers, some of the time. Customers replacing analog phone and fax systems with IP based systems need more. They need the same delivery rate and quality from the new systems — with all of their existing faxing options supported.

The FaxSIPit Network is a proven IP-based fax network that has transmitted millions of faxes successfully by insuring that all fax calls terminate to TDM lines — technology that has worked for decades. VoIP providers can route their local fax traffic to local TDM lines. Faxes transmitted outside the local area can be routed to the FaxSIPit Network.

A SIP Fax Trunking package from FaxSIPit is used in conjunction with FaxBack NET SatisFAXtion fax server software and includes IP based fax numbers (DIDs) that have been tested for fax use. Toll free DIDs are available with North America wide service. Local DIDs are available in many states and provinces.

## **Fonality, Inc. (402, OS 02)**

**P: 877-366-2548**

**[www.fonality.com](http://www.fonality.com)**

Fonality, [www.fonality.com](http://www.fonality.com), is a leader in business phone systems and contact center solutions for businesses of all sizes. Used by over 5,000 companies and 100,000 end users in 100 countries, Fonality's award winning IP-PBX VoIP phone systems have connected more than 225,000,000 mission critical phone calls. The PBXtra and trixbox Pro product lines are based on Fonality's patent-pending Anywhere Management™ Hybrid-Hosted™ architecture, plus an improved version of the popular open source Asterisk code base which has been modified to add reliability, stability and enterprise-class features. PBXtra and trixbox Pro deliver the advanced capabilities of an enterprise-class phone system for 40 to 80 percent less than traditional offerings. Fonality's fully free and open source telephony platform, trixbox CE ([www.trixbox.org](http://www.trixbox.org)), is home to one of the world's largest and fastest growing communities of open source telephony users, with over 200,000 live deployments and 125,000 new downloads each month. Fonality headquarters are in Los Angeles and company investors include Azure Capital Partners and Intel Capital.

## **FreedomVOICE Systems (634)**

**P / F: 800-498-2440 / 877-839-7892**

**[www.freedomIQ.com](http://www.freedomIQ.com)**

Simple to demonstrate and easy to install, FreedomIQ VOIP Hosted PBX brings in-demand technology to a vast small business market. FreedomIQ Partners enjoy a risk-free opportunity to generate recurring commission revenue and recruit other Partners to earn tiered commissions. We have no sales quotas and provide all client technical support.

## **Freeside Billing (OS 10)**

**P / F: 415-462-1624 / 866-491-2775**

**[www.freeside.biz/](http://www.freeside.biz/)**

Freeside is the premier open-source billing, trouble ticketing and provisioning software for VoIP and other online businesses. Freeside's open-source license frees you from licensing costs, proprietary databases and black-box applications. Freeside Internet Services offers turnkey billing and telephony service provider appliances, installation services and custom billing solutions.



A complete wholesale virtual fax solution  
from the nations leading Wholesale VoIP service provider.

Introducing teleFAX. A 100% turn-key virtual fax  
service designed to give your customers an affordable and  
easy to use fax solution. With features like quick fax,  
built-in text editor, multiple attachment support, over 8000+  
rate centers for local & Toll Free fax numbers, and  
API support...it's the perfect compliment to any VoIP  
service providers line card.

So why wait? Go to [www.telefax.com](http://www.telefax.com) or call  
888-898-4835 to sign up today.



quick fax

**Toll Free fax DIDs** built-in text editor

desktop client multiple attachments

white label **fax to email**

fax archiving fax DIDs in 8000+ rate centers  
**email to fax** dozens of supported  
file formats

Vitelity Communications  
6160 S. Syracuse Way, Suite 200  
Denver, CO 80111  
888-898-4835  
[sales@vitelity.com](mailto:sales@vitelity.com)

[www.vitelity.com](http://www.vitelity.com)

**Vitelity**  
Wholesale Carrier Services



# Packet8 Business Phone Service®



## Hosted PBX

The Packet8 Virtual Office hosted iPBX service eliminates the need for traditional business phone systems by delivering feature-rich telephony services over managed or unmanaged Internet connections. Businesses benefit from minimal start-up costs, reduced TCO and complete scalability.

## IP Trunking

With Packet8 SIP Trunking services, businesses gain the cost savings and flexibility associated with VoIP technology without having to purchase a new phone system or add hardware to their current setup.


## Hosted Call Center

Built from the ground up as a true hosted application, the Packet8 Complete Contact Center is a powerful, easy-to-use platform that ties together all customer communication channels, such as phone, email, web chat, and voice mail for single or distributed call center locations.

# A Solution for







I work from  
home and  
just need  
a single  
business line

I need a  
low-cost  
solution to  
replace my  
old key system

I make a lot  
of overseas  
phone calls,  
but the rates  
are killing me!

### Single Line Business

Ideal for SOHO professionals, Packet8 Business 2000 phone service is packed with features and priced for the cost conscious entrepreneur.

### Hosted Key System

Companies whose size or structure dictates the sharing of multiple, common phone lines among employees, regardless of where they are located, will enjoy the cost and performance advantages of the Packet8 Hosted Key System service.

### Mobile

Packet8 MobileTalk lets cell phone users reduce their international calling rates by as much as 90%. There are no special PIN codes to keep track of and no long access numbers to dial. MobileTalk provides the highest quality digital voice service and lowest calling rates to any international location.

# Every Business

With Packet8's expanded suite of business phone solutions, companies of any size, configuration or geographic location can benefit from the cost and performance advantages of VoIP technology with little or no upfront cost. Find the solution that best fits your needs at <http://Business.Packet8.net>.

**VISIT US AT**  
**BOOTH 407**  
**LEARN MORE**



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## G-Tek (336)

**P / F: 817-291-3300 / 817-251-9392**

**[www.GTek.com.tw](http://www.GTek.com.tw)**

G-Tek is a global corporation with primary focus on OEM/ODM/JDM telephony research and development, product design and manufacturing services for globally distributed, and partner-branded, VoIP telephony solutions. G-Tek VoIP partners are delivering an entire partner-branded SIP-based portfolio of VoIP and WiFi survivable gateways/PBXs, SIP single-/multi-line desk phones, and WiFi handsets.

## GlobalThink Technology SA (341)

**P / F: 54 351 5684789 / 1 786 999 0802**

**[www.globalthinktec.com](http://www.globalthinktec.com)**

GlobalThink is a leading vendor of N-Play products and solutions, for telecoms and service providers.

NGN Solution for N-Play Market  
Voice, Data, Video, GPS, etc.

Telco Billing, IPTV Billing, WiFi Billing, CAS, SIP-H323 Softswitch, Callshop, Softphone, IVR, Wholesale, Calling Card, IM, Conference, HPBX, IPPBX, CDRCapture, Provisioning, Reporting Server, BlueTooth Marketing.

## Gold Systems (148)

**P / F: 303-447-2774 / 303-447-0814**

**[www.goldsys.com](http://www.goldsys.com)**

Gold Systems develops and deploys voice-driven communications solutions utilizing IVR, text-to-speech and speech recognition. Our offerings include implementing unified communications and unified messaging telephony interfaces, customized contact center solutions, and password management for telephone-based self-service. Gold Systems is a top-tier partner to leading speech and messaging vendors, including Microsoft, Avaya, Cisco and Nuance.

## GotVoice, Inc. (541 & Mtg Rm 1)

**P / F: 425-216-3585 / 206-264-8221**

**[www.gotvoice.com/index.php](http://www.gotvoice.com/index.php)**

A pioneer in voice messaging services, GotVoice ([www.gotvoice.com](http://www.gotvoice.com)) enables the convergence of everyday voice into the digital world. Its patented communications platform empowers consumers and businesses to receive, consume and unify voice communications utilizing email clients, through their mobile phone interface or via the GotVoice Website. With popular features such as Voicemail-to-

Text, Visual Voicemail, the ability to share voice messages as digital audio files and the unification of wireless, landline, PBX and VoIP voicemail, GotVoice makes seamless communication possible, anywhere. Founded in 2003 by industry veterans with backgrounds from RealNetworks, Microsoft, Starwave, AT&T and Corbis, GotVoice works with all major mobile, business and residential carriers.

## Grande Communications (447)

**P / F: 512-878-4000 / 512-878-4286**

**[www.grandecom.com](http://www.grandecom.com)**

Grande Communications, Inc. provides rapid delivery of innovative wholesale broadband, carrier and ISP services. Enabled by its leading-edge optical network, Grande delivers communications solutions with flexible service options with a commitment to the customer's needs.

## Grandstream Networks (419)

**P / F: 617-566-9300 / 617-249-1987**

**[www.grandstream.com](http://www.grandstream.com)**

Grandstream Networks is a leading designer and manufacturer of innovative, affordable, and high quality IP voice and video products for the worldwide broadband telephony market. Our products are fully compatible with the SIP industry standard, field proven with large and rapidly growing deployed base, and have broad interoperability with the majority of 3rd party SIP products on the market today.

Grandstream continues to bring innovation to the IP communications market with exciting products of compelling values and differentiations. Grandstream Networks is headquartered in Brookline, Massachusetts with offices in Dallas, Los Angeles and Shenzhen/China.

## Gridborg America (448)

**P / F: 817-855-5160 / 817-522-5042**

**[www.gridborgamerica.com](http://www.gridborgamerica.com)**

Gridborg America does away with the need for expensive voice cards. One hardware server can support up to 1000 ports at a cost of less than \$20/port. There is no API to deal with. Any application can talk to the Gridborg server using simple text messages. This allows any programming language to communicate with the server quickly and easily. Reducing development time by as much as 60%. The Gridborg server is ideal for integrating into any software or hardware product. Communicating with simple text messages through TCP

the Gridborg Server can be easily collocated anywhere in the world. Eliminating bandwidth and recurring monthly charges for T1's, DS3, OC3's etc. Make your first VOIP phone call in less than 10 minutes using the Gridborg Server.

## i2Telecom (Mtg Rm 2)

**P / F: 404-567-4750 / 770-663-8282**

**[www.i2telecom.com](http://www.i2telecom.com)**

i2Telecom is an innovative developer of intellectual property and high value technologies in the communications and broadband space. i2Telecom provides high-quality international and domestic long distance calling services to subscribers at a fraction of the cost of traditional carriers by leveraging the power of the Internet. Our patented VoiceStick® and patent-pending InternetTalker™ access devices enable any telephone or business phone system (PBX) to access our global network and advanced routing technologies to complete most of the call over the Internet, paying only for the last leg of the connection. The Company's foundation in the Voice over Internet Protocol ("VoIP") has led to numerous product awards and industry wide recognition as an innovator with sweeping product and consumer acceptance for "ease of use" and practical applications.

Our patented VoiceStick™ device enables any telephone or business phone system (PBX) to access our global network and advanced routing technologies to complete most of the call over the Internet, paying only for the last leg of the connection.

Our newest product, MyGlobalTalk™, was just awarded Internet Telephony Magazine's "Product of the Year 2007" and Unified Communications® Magazine 2007 "Product of the Year Award". MyGlobalTalk™ leverages i2Telecom's® patented technology and previous Internet Telephony Magazine awards for "Product of the Year 2004" and "VoIP Service Provider of the Year 2005". In an industry "first", MyGlobalTalk™ places Internet telephony in the hands of every cell phone user, independent of wireless carrier technology, handset manufacturer or the type of wireless carrier voice/data plan involved. In addition, MyGlobalTalk™ is fully functional without local access to the Internet or proximity to an Internet "hotspot". Users also need not wait for the availability of a dual-mode WiFi phone,





Flexibility of  
open-source  
without the risk  
and uncertainty

Epygi can offer your customers the freedom of a true IP solution while keeping VoIP profitable for you.

If you want to select a VoIP solutions partner to help you grow your business, look no further than Epygi Technologies. Visit us at **Booth #434** to enter for the drawing of a FREE Quadro6L IP Key System.



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

because MyGlobalTalk™ provides the benefits of a dual-mode phone at a fraction of the cost using the custo

## **IgeaCare Systems Inc. (D7)**

**P / F: 905-707-1669 / 905-707-1775**

**[www.igeacare.com](http://www.igeacare.com)**

IgeaCare Systems Inc. is a company committed to manufacturing state-of-the-art telephony based communication systems.

We are dedicated to providing complete communication solutions for all spectrums of the Healthcare Market: Assisted Living/Retirement Homes, Nursing Homes/Long-Term Care Homes, Hospitals/Acute Care, Emergency Response and Home Care. IgeaCare Systems Inc. is an innovator in the field of intelligent communication solutions. Quality Care Through Innovative Technology...IgeaCare.

## **iKnowWare (225)**

**P / F: 512-633-8288 / 866-388-1652**

**[www.iKnowWare.com](http://www.iKnowWare.com)**

iKnowWare simplifies and unifies the way you work by connecting and communicating by a secure internet view, mobile phone, or VOIP PBX. iKnowWare makes your work easier as you have one single view of your customer, project, task, operations, marketing campaigns, inventory, and invoices/quotes. Certified Channel Partners make money and a reoccurring revenue by showing customers the way to reduce cost, make employees happy, and increase productivity/profits with easy to use Simple Subscription Solutions in 30 Days or less!! Roles based views/privileges on what YOU need to know based on device, location, group. Qualify as a Partner Today! Call 877-414-5669 or e-mail: Partners@iKnowWare.com

## **iLinc (626)**

**P / F: 800-767-9054 / 602-952-0544**

**[www.ilinc.com](http://www.ilinc.com)**

iLinc, a recognized leader in Web conferencing, desktop video conferencing software and collaboration solutions, aims to revolutionize the way organizations meet and communicate. Through its software and services, iLinc liberates people by enabling them to get more done, travel less, and achieve work-life balance while preserving the environment. iLinc offers the only enterprise-class Web and video conferencing software that allows customers to choose between a software-as-a-service (SaaS) rental model or a traditional software purchase model, in combination with hosting by iLinc or on-premise installation.

## **Infineon Technologies (MR5)**

**P / F: 866-951-9519 / 408-503-1588**

**[www.infineon.com](http://www.infineon.com)**

Infineon is a leading supplier of semiconductor solutions for carrier, enterprise and residential access applications. Ranked #1 in voice access chipsets, Infineon offers a comprehensive portfolio of TDM, VoIP and DECT telephony solutions for applications such as VoIP routers, DSL gateways, cable modems, WiMax CPE, FTTx, and PBX.

## **Ingate Systems (Mtg Rm 507)**

**P / F: 603-883-6569 / 413-460-0414**

**[www.ingate.com](http://www.ingate.com)**

Ingate® Systems develops firewall technology and products that enable SIP-based live communication for the enterprise while maintaining control and security at the network edge. Ingate has a long history of developing next-generation firewall technology that solves the NAT/firewall traversal issue with SIP communications. In addition to an extensive line of Ingate Firewall® products, the company also produces the award-winning Ingate SIPrator®, a device that connects to an existing network firewall to seamlessly enable SIP communications. Ingate products currently protect the networks of retail companies, financial institutions, industrial firms, government agencies and small-to-large enterprises throughout Europe, Asia and North America. Additionally, Ingate has established the SIP Trunking Network at [www.siptrunk.org](http://www.siptrunk.org), an educational resource that aggregates information from throughout the SIP trunking community in an industry-wide effort to educate enterprises about SIP trunking, to help simplify deployments.

Ingate Systems AB is headquartered in Sweden with offices in Stockholm and Linköping. Its wholly owned subsidiary, Ingate Systems Inc., is located in Hollis, New Hampshire.

## **IntelePeer (135)**

**P / F: 650-525-9200 / 650-287-2628**

**[www.intelepeer.com](http://www.intelepeer.com)**

IntelePeer, a provider of hosted communications services, enables software companies and telecom carriers to incorporate voice and multimedia capabilities to enhance their product offerings and drive new revenue streams. The company's open communications platform combines

a carrier-grade network with a rich development/runtime environment to reduce development time, customization, and interoperability testing.

## **Interactive Intelligence (514)**

**P / F: 317-872-3000 / 317-872-3000**

**[www.inin.com](http://www.inin.com)**

Interactive Intelligence Inc. (Nasdaq: ININ) is a global provider of unified business communications solutions for contact center automation, enterprise IP telephony, and enterprise messaging. In 1997 the company released the first "all-in-one" communications software suite offering single-platform architecture with inherent multi-channel processing minus the cost and complexity introduced by multi-point vendors. Interactive Intelligence is a leader in the adoption of voice over IP with its release in 2002 of the first converged communications application software suite to fully incorporate the session initiation protocol (SIP). Interactive Intelligence has combined this innovation with breadth of experience, today offering its more than 3,000 global customers a comprehensive solution-set comprised of premise-based and hosted offerings, including software, hardware, consulting, support, education and implementation. The company's long-term success is backed by hundreds of industry accolades, including Software Magazine's top 500 global software and services suppliers, NetworkWorld's top 200 North American networking vendors, BusinessWeek's top 50 hot growth companies, and FORTUNE Small Business Magazine's top 100 fastest growing companies. Interactive Intelligence was founded in 1994 and employs approximately 600 people. It's headquartered in Indianapolis, Indiana and has six global corporate offices with additional sales offices throughout North America, Europe, Middle East, Africa and Asia Pacific.

## **Intuitive Voice Technology (325)**

**P / F: 602-249-5750 / 602-249-5755**

**[www.IntuitiveVoice.com](http://www.IntuitiveVoice.com)**

Visit booth 325, where Intuitive Voice Technology will showcase its Evolution PBX, the industry's most advanced asterisk-based phone system for the SMB market. With the Evolution PBX, users enjoy functionality equal to the million-dollar PBX systems, but at a fraction of the cost. Come see why Evolution PBX has everyone talking! Reseller opportunities available. <http://www.IntuitiveVoice.com/>



# High Margins + Residual Revenue become a Partner today!

## SALES

iKnowWare to increase earnings... Do You?

Client Tracking and History

Quotes/Invoices

Customer Service

Opportunity Pipeline and  
Forecast

## MARKETING

ROI and Incentives

Lead/Demand Generation

Lead Tracking

Sales Promos / Cross Selling

Campaign Management

## FINANCE/OPERATIONS

View Real-Time ~

Quarter/FY Sales Revenue

Credit, Collect, Ship

Time line/Schedules

Inventory Control

Project/Task Status

## DISTRIBUTION

Software  
+ Services  
Easy to Use  
Sell/Support

## CUSTOMER



Call 877-414-5669 to register  
for Show Promo/Event Invite or  
e-mail: [Partners@iKnowWare.com](mailto:Partners@iKnowWare.com)

Visit us in Booth#225 ITEXPO @ LA Convention Center September 16th-18th





Ideas flourish. Decisions are made. And a hero is born. A network, collaborative. Communications merge mobility into a single conferencing work video work with e-together. Where the or wireless. And the story continues at



Decisions flow. Efficiency increases.  
(That's you.) On the integrated  
network, integration is effortless. Cisco Unified  
Integrates voice, video, data and  
security in a single secure system. So  
you can work with calendaring. Voice and  
email. And people can all work  
the way they want. When they want. Wired  
or wireless. That makes everyone look good. The  
Cisco Unified Communications System.  
[cisco.com/integratednetwork](http://cisco.com/integratednetwork).

welcome to  
the human network.





(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## **IR link corp. (252)**

**P / F: +82 2 404 3372 / +82 2 431 3056**  
[www.irlink.co.kr](http://www.irlink.co.kr)

IR link corp., located in Seoul Korea, has been specialized in supplying professional Call center solution based on self-developed headset phone system, so-called 'Ziphone' (Zi means wisdom in oriental character), now 4 models lined up to cover analog PSTN and IP network.

Lately IR link launched SIP headset phone, ZiphonX, designed to suit call center requirements, enabling the user to freely record and playback call conversation and work with any call center application program via ZiphoneX API.

ZiphonX, compact SIP headset phone, will provide Call Center clients and SI firms with most flexible, cost-effective solution to help them build up highly qualified IP based call center system.

## **Kentrox (248)**

**P / F: 503-643-1681 / 503-350-6110**  
[www.kentrox.com](http://www.kentrox.com)

Kentrox is a world-class innovator of site management solutions, QoS access routers and CSU/DSUs, enabling service providers and businesses to reduce network and operating costs while improving network performance. With more than one million products successfully deployed, Kentrox is a trusted partner of network operators and enterprises worldwide.

## **LAN Power Systems (326)**

**P / F: 866-674-0384 / 866-673-9083**  
[www.lan-power.com](http://www.lan-power.com)

LAN Power Systems makes 3rd generation Power over Ethernet (PoE) products available to the VoIP, Wireless and Security Systems marketplaces.

Flexibility, Simplicity and Value set LAN Power's line apart from other PoE solutions. The PoE solutions are used with existing non-PoE enabled Ethernet Switches saving bigtime \$\$ for VoIP, Wireless and Security Systems installations. LAN Power exceeds current IEEE802.3af PoE specifications making investment protection a reality. Today's most wanted features like support for Gigabit traffic, High Power availability on all ports simultaneously and full Modularity plus no required software are built into every LAN Power unit. A full 2 year warranty is included assuring peace of mind. Come see our

PoE products powering VoIP Phones, Wireless APs and IP Security Cameras in our Distribution Partner cyLogistics' booth # 326. Don't miss Special IT Expo Show offers from cyLogistics on LAN Power Systems PoE products available only for attendees of the Fall TMC IT Expo Conference.

## **Macquarie Equipment Finance (144)**

**P / F: 248-253-9000 / 248-339-1510**  
[macquarie.com/mef](http://macquarie.com/mef)

Macquarie Equipment Finance is a global provider of customized equipment leasing, asset finance, and asset management solutions and is a member of the Macquarie Group, a diversified international provider of financial, advisory, and investment services.

## **MARSHALL ELECTRONICS (347)**

**P / F: 310-333-0606 / 310-333-0688**  
[www.mxlomics.com](http://www.mxlomics.com)

Marshall Electronics produces microphones for every major studio and recording professional around the world. Marshall's focus at IT Expo demonstrate a new line of professional quality microphones products for price-conscious musicians, podcasters, digital videographers, and other creative digital enthusiasts. Our new line of conferencing and podcasting microphones start at \$99.

## **Matrix Telecom Pvt Ltd (641)**

**P / F: 714-706-9922 / +91 265 2636598**  
[www.MatrixTeleSol.com](http://www.MatrixTeleSol.com)

Matrix is an ISO 9001:2000 company into manufacturing of Telecom products like

- VOIP Phones
- VOIP ATA
- GSM FCT
- GSM Gateways
- PBX
- Hotel PBX

Products are manufactured as per international standards like FCC, CE, RoHS etc.

## **MERA Systems (241)**

**P / F: 800-858-2549 / 905-882-7414**  
[www.mera-systems.com](http://www.mera-systems.com)

MERA Systems develops industry-leading VoIP softswitches and session border controllers that enable VoIP carriers and service providers to manage VoIP sessions on their networks. The MERA solutions offer a powerful combination of efficiency, scalability and rich functionality, bringing added value and revenue-generating opportunities to carriers globally. To

date, MERA Systems is a key provider of highly reliable and cost-effective VoIP switching platforms and IP Centrex solutions. The company has 600+ deployments in 74 countries worldwide.

## **Message Technologies, Inc. (242)**

**P / F: 800-868-3684 / 770-240-7474**  
[www.messagegetech.com](http://www.messagegetech.com)

Message Technologies, Inc. (MTI) helps companies improve business performance by delivering IP-enabled hosted interactive voice response (IVR) and speech recognition solutions. Since 1982, our customers have counted on us for highly available and affordable Speech IVR design, development and hosted solutions, fully integrated with their entire call-center environment.

## **Microsoft (Mtg RM 501B-1)**

**P / F: 425-722-1858 / 425-936-7329**  
[www.microsoft.com/responsepoint](http://www.microsoft.com/responsepoint)

Microsoft Response Point is a complete, affordable, and hassle free phone system designed for small businesses up to 50 employees. Available from industry leading hardware vendors including D-Link, Syspine, and Astra, Response Point is easy to set up and manage, and lets you access phones, contacts, and features using only your voice, all with the push of a blue button.

## **Minuteman/Para Systems, Inc. (332)**

**P / F: 972-446-7363 / 972-446-9011**  
[www.minutemanups.com](http://www.minutemanups.com)

Para Systems, Inc. is a leading provider of power technologies including the Minuteman® uninterruptible power supplies (UPS) for computers, telecommunications systems, security systems, and network devices. The Minuteman® UPS products range from 400VA up to 24kVA. Para Systems also sells power distribution units, power management software and surge suppressors.

## **Mitel (302)**

**P / F: 602-231-5161 / 602-231-5163**  
[www.mitel.com](http://www.mitel.com)

Mitel delivers flexibility and simplicity in smart unified communications solutions and applications for organizations of all sizes. Combined with a full range of managed services that include voice and data network design and traffic provisioning, custom application development, and financial solutions packages, Mitel is reinventing how successful





## Come Visit PIKA at booth 102 and enter for a chance to win a PIKA WARP appliance!

PIKA WARP appliance is a development platform for Linux, Asterisk and other open source telephony applications.

- embedded processor
- optional FXS/FXO/BRI modules
- power fail transfer
- MOH / Paging
- Customized development environment
- 2 x 20 character programmable LCD display
- Flash memory



**Drop off your entry form at booth 102 to win a PIKA WARP appliance!**

Name:

Company:

Phone:

Email address:

Are you a:

☐ Linux Developer

☐ Asterisk Developer

☐ Open Source Developer (indicate platform)

Other (please explain)





(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

organizations gain competitive advantage by easily collaborating and communicating over distance and time with customers, colleagues and partners. Mitel is headquartered in Ottawa, Canada, with offices, partners, and resellers worldwide.

## **MobileMax (436)**

**P: 203-599-1925**

**[www.mobile-mx.com](http://www.mobile-mx.com)**

MobileMax (TASE: MBMX), founded in 2004, is a key player in the Fixed Mobile Convergence (FMC) arena, providing unique patent-pending unified communication solutions that enable users to have one device with one number, one address book and one voice mailbox, always using the lowest cost network for connectivity. The MobileMax technology-agnostic solution, licensed to service providers, carriers and IP/PBX providers, consists of the MobileMax Provisioning Server, a strong provisioning tool that generates MobileMax applications on the fly configured to clients' requirements and sends them to end user's mobile phones; and the Personal and Enterprise Unified Communication Mobile Client that reside in the mobile device. The solution easily integrates into any existing PBX system and supports over 500 mobile phone devices.

## **Multi-Tech Systems, Inc. (632)**

**P / F: 763-785-3500 / 763-785-9874**

**[www.multitech.com](http://www.multitech.com)**

Multi-Tech® Systems manufactures telephony, Internet, remote access, device networking and embedded communications products. Since 1970, Multi-Tech Systems, Inc., has designed and manufactured high-quality, innovative, global communication products and continues to provide exceptional service and support exceeding customer expectations. Today's

mission is the same featuring new technologies with an identical focus: connecting voice and data over IP networks. Multi-Tech products include: MultiVOIP voice/fax over IP gateways, FaxFinder network fax servers, RouteFinder Internet security appliances, SMSFinder turnkey SMS servers, MultiConnect external device servers, CallFinder cellular gateways, MultiModem cellular and PSTN device networking products, plus SocketModem embedded communications products.

## **NESSOCIET, INC. (132)**

**P / F: -15479 / -15482**

**[www.nessociet.net](http://www.nessociet.net)**

Nessociet is a leading Voice over IP (VoIP) software and related systems developing, marketing, and sales company. Nessociet has customers in Japan, China, North America, and India. Nessociet designs and develops its products in its corporate headquarters in Japan and software design center in Kolkata, India.

## **NetSapiens Inc (425)**

**P / F: 858-764- 5221 / 858-560- 8914**

**[www.netsapiens.com](http://www.netsapiens.com)**

NetSapiens simplifies the delivery of IP voice and multimedia business communication services by providing turnkey solutions. NetSapiens' SNAPsolution™ integrates applications software, hardware, and services that enable competitive providers to quickly deploy services for increased revenue. These applications range from pre-paid calling, Hosted PBX, to unified messaging, conferencing, and call center solutions.

## **NETXUSA, Inc (206)**

**P / F: 1-800-BUY-NETX / 864-233-4344**

**[www.netxusa.com](http://www.netxusa.com)**

NETXUSA, Inc. is a recognized leading distributor of Voice over Internet Protocol (VoIP) products and services. NETXUSA has a distribution channel of Independent resellers throughout the United States and foreign countries that provide the local end-user installation and support services.

NETXUSA has two distribution and sales centers located

in the United States. East and West coast facilities provide rapid regional response and economical product delivery to the dealer channel.

NETXUSA in-house certified engineers and an experienced customer service staff are available to service the dealer after the sale.

If IP enabling an existing legacy telephone system or planning a new VoIP PBX is your challenge, NETXUSA has a solution for you.

## **NMS Communications (CommDev 106)**

**P / F: 508-271-1000 / 508-271-1147**

**[www.nmscommunications.com](http://www.nmscommunications.com)**

NMS Communications provides enabling technology and tools for the rapid development and deployment of value-added services on mobile and converged networks. For both traditional and next generation applications, NMS platforms are used around the world as the foundation for a wide range of revenue-generating voice, video and data applications.

## **Octasic (MR 4)**

**P / F: 514-282-8858 / 514-282-7672**

**[www.octasic.com](http://www.octasic.com)**

Octasic Inc. is a global provider of media processing silicon and software solutions for the converged carrier, enterprise and end-point communication equipment markets. The company's leading quality VoIP DSP solutions are based on Opus, a unique asynchronous DSP architecture. Octasic allows next-generation equipment manufacturers to significantly reduce system costs by offering unmatched performance in terms of density and power consumption. Founded in 1998, Octasic is a privately-held company headquartered in Montreal, Canada.

## **OpenVox Communication Co, Ltd (437)**

**P / F: -82536130 / +1-815-550 1850**

**[www.openvox.com.cn](http://www.openvox.com.cn)**

Established in Shenzhen, China in 2002, OpenVox Communication Co., Ltd is dedicated its passion in supplying Open Source Computer Telephony hardware and software products. With the people's expertise in design and service experience, we provide professional quality products and a 3-month "no questions asked" return policy for all OpenVox hardware.

All our hardware products are ready to be used with leading open source telephony





Win A Brand New Toyota Prius – Visit us at booth 626



**\$4/Gallon? - Ouch!**

## Meet online. Get more done.

Why suffer the costs, productivity losses, and environmental impact of driving or flying when you can meet clients and prospects online using iLinc web and desktop video conferencing. Starting at just \$49/user per month, it's a no brainer. Really.

Visit us at [www.iLinc.com/dailydemo](http://www.iLinc.com/dailydemo) to see for yourself and start saving valuable time—and use that gas money for something else.

**602-952-1200**

iLinc Communications, Inc.  
2999 N 44th St, Suite 650  
Phoenix, AZ 85018

**iLinc**<sup>TM</sup>  
Better Than Being There.



# INTRODUCING HUD 3

## THE DEFINITION OF UNIFIED COMMUNICATIONS™

- + COMMUNCIATE SMARTER  
CALL, CHAT, AND SMS WITH A SINGLE CLICK
- + WORK MORE EFFICIENTLY  
PRESENCE MANAGEMENT & DETECTION
- + RUN YOUR BUSINESS FROM ANYWHERE  
WORK FROM HOME, OFFICE, OR THE ROAD



## AWARD WINNING IP - PBX

**PBXtra®**  
by Fonality

- + UP TO 20% MARGINS
- + SOFTWARE & HARDWARE
- + FULL TURNKEY SOLUTION



**FONALITY**  
communicate openly



- + DEMOS EVERY HOUR
- + FREE T-SHIRT!

VISIT  
BOOTH  
#402



SOLUTIONS FOR RESELLERS

**trixbox**®  
by Fonality

- + UP TO 50% MARGINS
- + SOFTWARE ONLY
- + FULLY BRANDABLE





(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

projects including Asterisk®, trixbox®, Elastix® ISDN4BSD, AskoziaPBX, etc. OpenVox product line includes PCland PCI-e versions of T1/E1; PRI/BRI/FXO/FXS VOIP voice cards as well as hardware echo cancellation products. All are of excellent performance and being continuously extended with innovation.

## OrecX (OS 05)

**P / F: 312-895-5292 / 312-895-5220**

**[www.orecx.com](http://www.orecx.com)**

Based in Chicago, OrecX provides open source VoIP recording solutions at a fraction of the cost of proprietary recording applications that run \$1,000--\$4,000 per user. OrecX is the primary developer and sponsor of the Oreka open source call recording project hosted on sourceforge.net - [www.oreka.org](http://www.oreka.org). Over 30,000 users have downloaded the open source version of Oreka.

## Packet Island, Inc. (338)

**P: 408-329-9604**

**[www.packetisland.com](http://www.packetisland.com)**

Founded in 2004, Packet Island Inc. is a privately-held, venture-backed company located in Cupertino, California. A leading innovator of Voice-Data Network Management solutions, Packet Island has pioneered the development of award-winning SaaS-based solutions for managing converged media networks for SMB and enterprise multi-site networks.

## Pactolus Communications Software Corp. (319)

**P / F: 508-616-0900 / 508-281-5698**

**[www.pactolus.com](http://www.pactolus.com)**

Pactolus Communications Software ([www.pactolus.com](http://www.pactolus.com)) is the leading developer of service provider hosted SIP-based services processing 3-4 billion Minutes-of-Use monthly in 140+ service provider NGN networks worldwide. Customizable turnkey services include Operator-assisted Large Event and Reservation-less Audio Conferencing, Residential VoIP, Business VoIP – hosted IP PBX with voice mail, Prepaid Residential / Business VoIP, and Prepaid / Post Paid Calling Card.

## PAETEC (441)

**P / F: 866-945-3725 / 585-000-0000**

**[www.paetec.com](http://www.paetec.com)**

PAETEC is personalizing communications solutions for business customers across the United States. We offer a comprehensive suite of data, voice, and IP services, as well

as enterprise communications management software, network security solutions, CPE, and managed services.

## Pangea Communications Corp. (633)

**P / F: 503-221-1111 / 503-221-3080**

**[www.pangea-comm.com](http://www.pangea-comm.com)**

Pangea provides a full suite of Internet fax services and solutions. Our Virtual Fax Reseller program is a hosted fax service designed specifically for Telcos and VOIP companies to offer their clients a private-branded fax solution.

## PhoenixSoft, Inc. (342)

**P / F: 925-373-9100 / 925-373-9200**

**[www.phoenixsoft.com](http://www.phoenixsoft.com)**

PhoenixSoft develops the Cirrus Softswitch, including an integrated IP Transcoder and Session Border Controller to support both traditional and SIP-based VoIP switching. Features supported include Class 5 and Class 4 switching, Web Retail Portal, Prepaid and Postpaid calling, IN, IP PBX, Conference Bridge. Quality solutions with outstanding customer support since 1985. Turnkey and Hosted solutions available.

## PhoneFusion (244)

**P / F: 954-607-4400 / 954-607-4400**

**[www.phonefusion.com](http://www.phonefusion.com)**

PhoneFusion™ is a leading provider of reliable communications solutions to businesses of all sizes. The company's portfolio of Unified Communications and Fixed Mobile Convergence products combines all modes of communication – from mobile phones to traditional landlines to VoIP to Business Intelligence Suites and Distributed Call Centers – into one, single, manageable source. With PhoneFusion, organizations can confidently and seamlessly manage their daily business communications from any location, on any device, at any time.

The company offers three distinctive business communications solutions, designed to meet the needs of small-to medium-sized businesses and resellers alike. All the PhoneFusion products are built on cutting-edge technology with innovative and reliable solutions to combat today's common business communications problems. From home offices to the virtual workplace, PhoneFusion's portfolio of business communications solutions enables organizations to stay connected with sales, customers and prospects whenever and wherever they are.

## Phonevite (344)

**P / F: 408-907-5670 / 408-521-0392**

**[www.phonevite.com](http://www.phonevite.com)**

Phonevite is the new, safe and fun way of reaching your groups instantly over the phone. The service uses the latest in VoIP and web technologies to make it possible for anyone to get in touch with groups of people instantly over the universal device, the phone.

## Plantronics (643)

**P / F: 866-233-6878 / 916-357-5193**

**[www.plantronics.com](http://www.plantronics.com)**

Plantronics, Inc. introduced the first lightweight communications headset in 1962 and is today the world's leading designer, manufacturer and marketer of lightweight communications headset products.

Plantronics headsets are widely used in many Fortune 500 corporations and have been featured in numerous films and high profile events, including the historic "One small step for man" transmission from the moon in 1969.

Plantronics offers mobile headsets to address the cordless and mobile phone market, next-generation computer audio headset products for computer applications and corded and cordless headsets and systems for the office, small office/home office and contact centers.

## PortaOne, Inc. (431)

**P / F: 866-747-8647 / 866-747-8647**

**[www.portaone.com](http://www.portaone.com)**

PortaOne offers complete and scalable solutions for VOIP service providers.

As a leading global vendor of carrier grade software, and with customers in over 200 countries, we enable ITSPs to run a broad line of telecommunication services.

## Pulse, Inc. (534)

**P / F: 508-660-0340 / 508-660-0339**

**[www.pulsewan.com](http://www.pulsewan.com)**

Pulse, Inc. is a national VAR with solutions and products for VoIP, Wireless, IP and Ethernet Extension, Signaling and Interface conversion and traditional wide area networking. Pulse also offers Talking Platforms, a hosted PBX and ITSP white label service for resellers and service providers. Pulse has been in business for over 12 years and maintains 6 offices across the USA.



**NET's and Quintum's Intelligent  
VoIP Switches and Gateways  
are *The Perfect Fit* for  
Microsoft Office  
Communications Server 2007**

NET and Quintum, Microsoft Gold Certified Partners, unify VoIP applications and traditional telephony. And now, our VoIP solutions allow easy integration of Microsoft Office Communications Server 2007 with your existing telephony communications network.

The VX Series and Tenors offer real-time, seamless communication between Microsoft Office Communications Server 2007 and the legacy world. Their intelligent 'MultiPath' switching design assures the easiest deployment and minimum disruption to the existing communications network.

Quintum Tenors feature an intelligent **Unified Communications Proxy (UCP)** which provides 'Any to Any' connectivity between the Microsoft UC cloud and any existing network, including the PSTN, TDM equipment and SIP-based VoIP equipment. Check out our animated demo at [www.quintum.com/microsoft](http://www.quintum.com/microsoft)



# Traditional VoIP gateways just aren't the best fit for Unified Communications



Have questions on deploying  
Microsoft OCS2007?  
Ask Dr.Q at [drquintum@quintum.com](mailto:drquintum@quintum.com)



The Perfect Fit

[www.net.com](http://www.net.com)

[www.quintum.com](http://www.quintum.com)





(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

## Quanta Syspine (548)

**P / F: 877-427-9774 / 303-465-0237**

**[www.syspine.com](http://www.syspine.com)**

Quanta Computer, a Fortune Global 500 Company, has designed the award-winning Syspine™ Digital Operator Phone System an advanced IP PBX system that features Microsoft® Response Point™ phone system software specifically for companies with one to 50 employees. Syspine features a powerful voice-recognition system that offers SIP trunking for external VoIP calling, DID number, call history view, click to call and call restriction per phone. Syspine can be linked with a company's internal phone directory as well as an individual's Microsoft Office Outlook® address book. The system was designed for small businesses and is highly intuitive for easy learning and operation and can be set up in a matter of minutes. The Syspine Digital Operator A50 is a single solution device that will connect up to eight standard telephone lines; it features a high-end, server-grade, fan-less design and can be installed anywhere. Businesses can add up to 50 individual Syspine IP310 business phones to create a complete, integrated small business phone system with no additional licensing fees. To learn how you can become a Syspine VAR please visit us at [www.syspine.com](http://www.syspine.com).

## Quintum Technologies and Network Equipment Technologies, Inc. (420)

**P / F: 732-460-9000 / 732-544-9119**

**[www.quintum.com](http://www.quintum.com)**

Quintum Technologies, LLC, a wholly owned subsidiary of Network Equipment Technologies, Inc., manufactures Tenor VoIP MultiPath Switches and Gateways which support analog and digital communications in enterprise and service provider networks, offering H.323 and SIP-based support for IP telephony migration, legacy VoIP enablement, multi-office VoIP, survivable branch office, Hosted IP PBX, SIP trunking, and unified communications applications.

For nearly a quarter of a century, Network Equipment Technologies, Inc. (NYSE: NWK), headquartered in Fremont, CA, has provided voice and data communications equipment for enterprise and government multi-service networks requiring high performance and reliability. NET's VX Series is a line of feature rich, high performance VoIP switching gateways. The VX offers

VoIP solutions designed to support the integration of unified communications into the existing Voice TDM and VoIP network environments. Supporting robust security features, high call processing rates, and active directory, the VX can support the most demanding UC deployments.

Both Quintum and NET are Microsoft Gold Certified Partners. Quintum is a Nortel Developer Partner and an Avaya DeveloperConnection Partner, and NET is an HP Business Partner.

## Qwest Communications (426)

**P / F: 800-315-2000 / 210-541-3069**

**[www.qwest.com/wholesale](http://www.qwest.com/wholesale)**

Qwest offers a unique and powerful combination of voice and data solutions. Customers worldwide are turning to Qwest's industry-leading fiber optics and its spirit of service for quality products and superior customer experience. Qwest wholesale services: competitive local exchange carriers, enhanced service providers, incumbent local exchange carriers, international service providers, internet service providers, resellers, cable service providers and wireless service providers.

## Redfone Communications LLC (OS 09)

**P / F: 786-544-1196 / 786-544-1110**

**[www.red-fone.com](http://www.red-fone.com)**

Redfone is a hardware and software engineering firm dedicated to providing solutions for the most demanding Open Source based voice implementations in the World.

## Ring Carrier (326)

**P / F: 650-694-4949 / 650-694-4953**

**[www.ringcarrier.com](http://www.ringcarrier.com)**

Ring Provider™ is a complete VoIP solution for your network. Ring Provider gives you everything you need to immediately launch your private labeled VoIP business with low CAPEX and OPEX. Ring Provider is a complete, integrated, turnkey, VoIP, video and Web services solution. Ring Provider is built with open source and open standards. Open source means Ring Provider is a fraction of the cost of competing systems. Ring Provider offers you unprecedented revenue generating opportunities.

Ring Provider easily scales to unlimited numbers of enterprises and subscribers.

Start with a PBX Building Block™. A PBX Building block is a stand alone PBX. You

can install a PBX building block stand alone on a server at a customer site. Combine PBX building blocks and use multiple servers for greater performance and reliability.

Host PBX building blocks on the Ring Provider VoIP Fabric™. The VoIP Fabric lets you host an unlimited number of PBX building blocks. Combine hosted PBX building blocks to provide greater capacity, reliability and performance for your customers. The PBX Fabric maximizes performance and minimizes your hardware costs.

## RNK Communications (625)

**P / F: 781-613-6000 / 781-297-2091**

**[www.rnkcom.com](http://www.rnkcom.com)**

Established as a facilities-based carrier in 1997, RNK Communications™ is a full-service communications provider. RNK's unmatched operational efficiency, industry expertise, and technical innovation, enable it to offer a variety of customized communications products and solutions. For additional information call 888-642-9831 or visit [www.RNKcom.com](http://www.RNKcom.com).

## SANGOMA (519)

**P / F: 905-474-1990 / 905-474-9223**

**[www.sangoma.com](http://www.sangoma.com)**

Sangoma is a premium provider of PC-based telephony hardware and software products for proprietary and open-source-based networking and telephony solutions.

Sangoma's telephony and data hardware empower your PC to deliver superior auditory quality when communicating with public telephone networks.

With Sangoma, you're not limited by your choice of application: Asterisk®, FreeSwitch™, CallWeaver™, Yate™. Incorporate our hardware into a high-quality, cost-effective PBX or simply route data. Sangoma – unparalleled price and performance.

## Sansay (220)

**P: 858-678-0200**

**[www.sansay.com](http://www.sansay.com)**

Sansay is an innovative developer of advanced carrier VOIP infrastructure solutions. The company's products include VOIP and video session control, call routing, NAT/firewall/security and protocol adaptation/translation solutions. Sansay's comprehensive solutions allow customers to create "open but secure"



# The World's Communications Conference

The industry's best-attended event attracting thousands of Communications Professionals seeking new partners and the latest information



Diamond Sponsor:



Supported By:



## Mark Your Calendar!

February 2-4, 2009



## Join Us in Miami for ITEXPO East 2009

Reserve Your Space Now for the First  
and Biggest Event of 2009 - ITEXPO!



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

connections between their networks and those of their VOIP carrier partners, and to allow "open but secure" access to by enterprise and residential subscribers via public Internet and/or VPN.

Sansay's key executives were founders and founding members of Nuera Communications where, from 1995-2002, they led teams responsible for product management, engineering, marketing, strategy and business development.

## **Seawolf Technologies, Inc. (631)**

**P / F: 516-393-2150 / 516-393-2153**

**[www.seawolftech.com](http://www.seawolftech.com)**

Seawolf Technologies, Inc. is a leading Telecommunication and Internet Service provider offering complete voice and data solutions for the next generation networks. Highly integrated software, hardware and termination solutions enable their customers to capitalize on revenue opportunities in the convergence market of voice and data services by accelerating time-to-market and reducing their development and operational cost.

With their business locations in both the US and China, they focus on the world's two largest communication markets, offering a wide range of telecommunication services, such as online billing and prepaid calling card platforms, voice over IP (VoIP) solutions, global termination, international call forwarding and callback, domestic and international long distance service, nationwide 800/888/toll free service.

Their mission statement is, "We bring new technologies into your business." They are committed to keeping customers on the cutting edge of technology.

## **SimpleSignal (642)**

**P / F: 800-873-7670 / 866-456-4720**

**[www.simplesignal.com](http://www.simplesignal.com)**

SimpleSignal is a next-gen telco service provider that allows developers to integrate voice features into Web applications, such as Microsoft hosted exchange, Salesforce.com, Facebook, and the iPhone. Our resellers benefit by being able to sell the latest innovations in cloud based communications to SMB's worldwide.

## **SIP PRINT LLC (444)**

**P / F: 888-888-4108 x107 / 888-888-4028**

**[www.sipprint.com](http://www.sipprint.com)**

SIP PRINT is the worlds best VOIP/SIP based voice recording system.

## **SiTel Semiconductor (MR3)**

**P / F: 1-888-SITEL-SC / 408-432-7235**

**[www.sitelsemi.com](http://www.sitelsemi.com)**

SiTel Semiconductor BV was established in June 2005, following a management buy-out from National Semiconductor Corporation. Headquartered in 's-Hertogenbosch, The Netherlands, SiTel is a fabless semiconductor company with design centers in The Netherlands and Greece along with regional sales offices in Hong Kong, Japan and the U.S. The company specializes in the design, manufacturing and marketing of high-performance CMOS solutions for digital cordless and corded communication and is the world's number one DCT supplier for cordless voice, real-time data and VoIP.

## **snom technology AG (651)**

**P / F: 978-686-1531 / 978-686-1797**

**[www.snom.com](http://www.snom.com)**

snom technology AG develops and manufactures Voice-over-IP (VoIP) telephones based on open standard for enterprise communications.

snom phones can be operated with a large number of compatible IP-PBX that support the SIP (Session Initiation Protocol), giving customers a great choice for enterprise communication systems.

Recognized for its high-quality, customizable and cost-effective business solutions, snom is also differentiated by the company's ten-year history in the VoIP industry, and its dedication to high security standards.

The devices are suitable for use in all business environments ranging from home offices to small- and medium-sized enterprises and large corporations. snom also works directly with carriers, Internet Service Providers, and OEM customers. The company is globally present through branch offices and a distribution partner network.

## **SoTel IP Services (133)**

**P / F: 314-787-1800 / 314-787-1769**

**[www.sotelips.net](http://www.sotelips.net)**

SoTel IP Services is an ITSP offering a full suite of VoIP products for SMB and Enterprise customers through agent resources.

## **Spenser Communications, Inc. (544)**

**P / F: 800-454-4442 / 626-593-3232**

**[www.spensercommunications.com](http://www.spensercommunications.com)**

Telecommunications provider

## **Spracht (647)**

**P / F: 918-294-0036 / 650-318-8060**

**[www.spracht.net](http://www.spracht.net)**

Spracht is the acoustic division of parent company ODI (<http://www.odi.net>), a recognized leader in digital imaging, acoustics, LCD image display, and other consumer electronics. Founded in 1993, Silicon Valley based ODI has developed bestselling products, such as the Connectix Quickcam, for both Fortune 100 companies and technology notables such as Palm and Kensington. Spracht is currently positioning the Aura Mobile BT and Aura SoHo for businesses with conferencing needs.

## **Stratus Technologies (547)**

**P / F: 978-461-7619 / 978-461-3620**

**[www.stratustelecom.com](http://www.stratustelecom.com)**

Stratus is a provider of converged VOIP solutions and services for telecommunication service providers. By helping its customers connect their world using innovative solutions that they can trust and backing it with people they can count on, the Stratus Technologies Telecommunications Group is making convergence simple. Stratus Technologies solutions are installed in 14 of 20 of the world's largest carriers. Stratus has 27 years of expertise in server and services technology and has been a trusted solutions provider to customers in telecommunications, manufacturing, life sciences, financial services, public safety, transportation & logistics and other industries. For more information, visit [www.stratustelecom.com](http://www.stratustelecom.com).

## **Synway (ComDev 117)**

**P: 8.6571888606e+012**

**F: 8.6571888509e+012**

**[www.synway.net](http://www.synway.net)**

Synway specializes in designing hardware/software building blocks for use in Computer Telephony Integration (CTI) applications, such as IVR, Call Center, Recording, Unified Messaging and Value-Added Service (VAS) in both PSTN and IP environments. Our products feature rich media processing resources including Fax, Conferencing, Codecs, Echo cancellation and call control with an array of signaling capability for SIP, SS7 packets, ISDN and CAS in worldwide IP/T1/E1/Analog networks.

In the past two decades, Synway has attracted CTI solution providers through superior service, field-proven products and matchless pricing for the highly reli-



# Everything You Need to Know About SIP Trunking

Get the facts about SIP trunking. Learn everything you need to know from the industry's leading experts in these educational seminars, free for all ITEXPO attendees.

## Live Demos

Watch as a SIP trunk is deployed live — on-site — in 20 minutes or less.

Tuesday, September 16, 2008

SIP Trunking Professional Development Program  
- enterprise infrastructure      - service provider view

Wednesday, Sept 17, 2008

The Mechanics of SIP Trunking  
- implementation    - security    - solutions    - case studies

Thursday, Sept 18, 2008

SIP Trunk "Basic Training"  
- Avaya    - Ingate

Meet us at the Los Angeles Convention Center

Room 507

More information: [www.ingate.com/SIP\\_Trunking\\_seminar.php](http://www.ingate.com/SIP_Trunking_seminar.php)

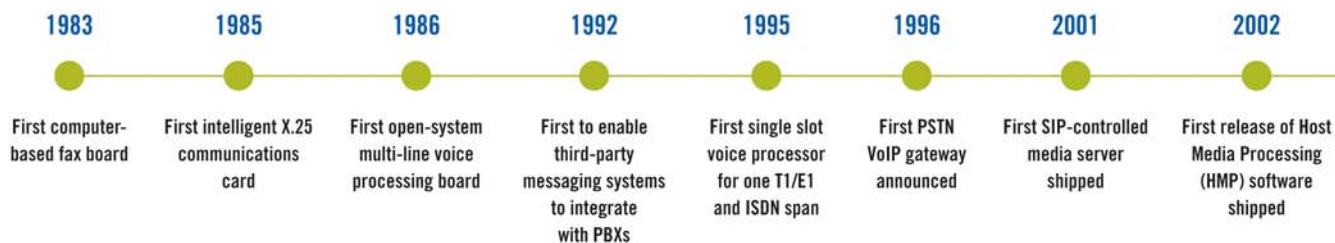




## On A Global Stage, Making Innovation Thrive

Whenever technology has led developers to a crossroads, Dialogic has been there, working with its partners to ensure they had the tools to make the transition to new and innovative applications. In addition to its storied history in the development and proliferation of advanced computer telephony solutions, Dialogic has been at the forefront of every technological wave, including Voice over IP (VoIP), fax over IP, SS7 integration, and most recently the integration of video with traditional voice.

Dialogic maintains over 50 patents and through its participation in standards bodies and organizations such as the IETF, Dialogic has helped craft industry standards for solutions such as the Media Resource Function (MRF) in 3GPP's IP Multimedia Subsystem (IMS). Dialogic also created the first SIP media server.



**SEE US AT TMC UNIVERSITY:** "SIP: The Foundation for Building Enhanced Services."

Thursday, September 18, 2008. 8:15am – 4:00pm; Room 510.

For more information, visit: [www.tmcnet.com/voip/conference/west-08/tmc-university-sip.htm](http://www.tmcnet.com/voip/conference/west-08/tmc-university-sip.htm)



# 45 million ports deployed



## Build on Dialogic

Dialogic is a catalyst for innovation in communications

- Market Leader in multimedia-enabling technologies for building innovative Value-Added Services and Messaging solutions
- Rich selection of open-standards components for IP, Mobile, and Video
- More than 45 million ports shipped worldwide
- Global sales and support with 35+ offices



Dialogic® Multimedia Kit for PCIe



Dialogic® 4000 Media Gateway Series



Dialogic® Host Media Processing Software & Dialogic® HMP Interface Board



Dialogic® Brooktrout® TR1034 Fax Board & Dialogic® Brooktrout® SR140 Fax Software

2003

First commercially available SS7 PCI board to exceed 20K MSU/s

2004

Media server platform for first 3G video messaging application deployment

2006

First IP-based fax at V.34 speeds shipped

2007

First IP multimedia processing board (AdvancedTCA) shipped

2007

First "hybrid" gateway shipped for Microsoft® Office Communications Server 2007 environment with integrated Mediation Server

2007

First 3G-324M interface shipped on an AdvancedTCA board

2008

First Fax over IP shipped with V.34 and T.38 support

[www.dialogic.com](http://www.dialogic.com)



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

able Telco and enterprise communications platform. With internally coordinated Maximum Support System (MSS), our service engineers provide pre-sales consulting, development support, and after-sales service, so you can monitor the service process effortlessly. Synway's competitive pricing will ensure your matchless competitiveness in any project bidding or distribution-channel expansion globally. Our products have been applied by hundreds of customers worldwide.

## **Taqua LLC (637)**

**P / F: 972-692-1800 / 972-437-2762**

**[www.taqua.com](http://www.taqua.com)**

Taqua's switching and enhanced applications are deployed by wireline, cable and wireless carriers as next-generation end-office exchanges with enhanced IP and Web 2.0 communications solutions. Taqua's lowers switching deployment and operational costs, while delivering the world's most advanced new IP-based services.

## **Telcobridges (ComDev 619)**

**P / F: 450-655-8993 / 450-655-9511**

**[www.telcobridges.com](http://www.telcobridges.com)**

TelcoBridges provides the best technologies telecom solution developers can use to realize their bright ideas in order to create a clear future.

Our customers are telecom solution developers building carrier-grade VoIP or TDM solutions for the media gateway, switching and value-added services markets.

By supplying the industry's most pioneering hardware and software products, TelcoBridges is on its way to become the preferred supplier of telecom platforms. Founded in 2002, TelcoBridges and its customers have deployed systems in more than 35 countries for some of the world's largest telecom operators.

Headquartered in Montreal, Canada, where TelcoBridges performs its R&D and production activities, the company has further sales representation and technical support staff in three regional offices: USA (San Jose, CA), China (Hong Kong and Beijing), and Korea (Seoul).

Our expertise is carrier-grade telecom platforms for media gateway, switching and VAS solutions. Our products provide

industry-leading architecture, flexibility and scalability. Our platform delivers the industry's fastest time-to-market and a vital competitive edge. Our customers are developers building carrier-grade VoIP and TDM solutions.

## **TeleMatrix / Scitec (443)**

**P / F: 719-638-8821 / 719-638-8815**

**[www.telematrixusa.com](http://www.telematrixusa.com)**

A Leader in Hospitality and Business Telephone manufacturing. VoIP, Digital and Analog sets available.

The 3300IP corded & 9600IP cordless hospitality, and IP550 & IP100 business terminals are SIP compatible and reasonably priced.

## **Tigerpaw Software (648)**

**P / F: 402-592-4544 / 402-592-7317**

**[tigerpawsoftware.com](http://tigerpawsoftware.com)**

Tigerpaw is an award-winning, all-in-one business software system that seamlessly integrates to popular accounting programs. Tigerpaw helps converged solution providers successfully streamline and manage their quoting, sales, marketing, installation, technicians, service and administrative functions such as purchasing, invoicing, inventory. All Tigerpaw functions integrate mobility, web and GPS technology, arming Telecom and IT Managers with the tools to improve customer interactions, bill faster and streamline operations top to bottom.

## **Time Warner Cable Business Class (414)**

**P: 703-345-3642**

**[www.twcbc.com](http://www.twcbc.com)**

Time Warner Cable Business Class, a division of Time Warner Cable (NYSE: TWC), offers a full complement of business communications tools to small and medium-sized businesses and enterprise-sized companies. Its data, voice, video and security solutions are enhanced by award-winning customer service and local support teams.

## **Touchstone Technologies, Inc. (526)**

**P / F: 215-672-6550 / 215-672-6551**

**[www.touchstone-inc.com](http://www.touchstone-inc.com)**

Touchstone Technologies specializes in delivering state-of-the-art solutions that enable our customers to bring their products and services to market with confidence. Leveraging our diverse customer base, our products assure interoperability with a wide array of industry-leading voice and video applications and equipment.

Our powerful, scalable, and affordable solutions are the ideal set of tools for developers, engineers, integrators, quality assurance specialists, and network administrators. As one of test and measurements earliest pioneers, Touchstone has been proudly serving the communications industry since 1989.

## **tw telecom (552)**

**P / F: 213-929-2100 / 213-929-2160**

**[www.twtelecom.com](http://www.twtelecom.com)**

tw telecom delivers the most advanced network provider solutions on the market to a wide spectrum of businesses and organizations across the US. Since 1993, we have built a legacy of success through exceptional customer care, remarkable products and powerful networks. These strengths enable us to deliver scalable solutions that help customers improve their business metrics.

## **U4EA Technologies (542)**

**P / F: 15102793355 / 15106870006**

**[www.u4eatech.com](http://www.u4eatech.com)**

U4EA Technologies is a leading provider of Multi-Service Business Gateways that enable service providers to deploy highly integrated, secure, and wireless IP communications solutions to SMB customers. U4EA's all-in-one CPE devices, supported by the company's patented QoS technology, ensure the optimal delivery of converged voice, data and video services.

## **Vitelity Communications - Wholesale Carrier Services (219)**

**P / F: 888-898-4835 / 303-991-7977**

**[www.vitelity.com](http://www.vitelity.com)**

Vitelity Communications is a rapidly growing leader in Wholesale VoIP services. Whether you are servicing SMB or residential, we have the products and services your customers want, including: Origination, Global Termination, DIDs (8,000 rate centers), Toll Free, vFAX, Leased PBX, e911, etc. Robust API and user portals are also available.

## **VIXXI Solutions, Inc. (536)**

**P / F: 972-791-0911 / 972-717-9099**

**[www.vixxisolutions.com](http://www.vixxisolutions.com)**

Completed and launched in the early 2nd quarter of 2007, VIXXI's national IP-based E911 solution, the VIXXI Geospatial E911 Routing System™, truly represents a processing breakthrough for the emergency network system. Composed of the



# VOCALCOM

CRM & Call Center Technology

Inbound ACD & IVR • Outbound Predictive Dialing • Virtual Hosted Solutions • Agent Scripting Tools  
Avaya CTI & Dialer • Web-based & MultiChannel Solutions

## VoIP CONTACT CENTER SOLUTIONS

### Solutions Provider For...

- Customer Service
  - Collections
  - Telemarketing
- Outsourced Call Centers
- Communications and Media
- Education & Student Loans
- Financial Services & Mortgages
  - Fundraising
- Insurance Services
  - Travel & Resorts
  - Healthcare



A Global Approach to  
Call Center Technology



Tel: 188VOCALCOM (1 888 622 5266) • email: [world-sales@vocalcom.com](mailto:world-sales@vocalcom.com) • [www.vocalcom.com](http://www.vocalcom.com)

Argentina • Belgium • Brazil • Canada • Denmark • France • Germany • Italy • UAE  
Morocco • Mexico • Netherlands • Portugal • Spain • Tunisia • UK • USA



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

EG200™ and VIXXI-Link™ service, this solution meets the recently passed U.S. House 911 Modernization and Safety Act.

The VIXXI EG200 arrives at the PSAP location completely configured requiring only a broadband connection, and a few minutes of time. VIXXI-Link includes a web portal for end user registration, total ALI management with MSAG validation and ERDB functions, and emergency call routing.

## **VOCALCOM (226)**

**P / F: 514-733-6444 / 514-733-7999**

**[www.vocalcom.com](http://www.vocalcom.com)**

Over the past 10 years, Vocalcom has provided world-class contact center solutions. With a global presence in 37 countries and over 2500 clients using our systems daily, Vocalcom occupies a leadership position in today's call center marketplace.

Within the Vocalcom product portfolio list you will find the latest ACD, IVR, PREDICTIVE DIALING, HOSTED SOLUTIONS, CALL RECORDING, CTI, SCRIPTING SCREEN POP, SPEECH ENGINE, HOSTED SOLUTIONS, VOIP/HMP ENGINE, REPORTING, SUPERVISION AND EMAIL MANAGEMENT. Together these next generation technologies can be bundled with Vocalcom's software architecture to provide turn key solution for all industry segments. Hermes addresses itself to all industry sectors that directly target customer interactions: sales, marketing, collections, customer services and telemarketing.

Hermes.Net is an integrated, full-featured multi-channel contact center solution based on a single IT platform. The major advantage of this unique architecture is that it combines all separate contact center elements. It is a flexible platform specifically designed to be integrated with your existing infrastructure and IT applications. Hermes.Net is a completely browser-based platform enabling you to access all contact center functionalities through a standard Internet browser.

All of our office locations, and throughout our development and sales operations, Vocalcom maintains a deep commitment to our customers. Our solutions have allowed our customers to efficiently service their clients by using advanced feature, rich functionality and superior stability.

## **VoIP Logic (435)**

**P / F: -5288 / -7901**

**[www.voiplogic.com](http://www.voiplogic.com)**

VoIP Logic is a leading global provider of VoIP Managed Services and Solutions. The company enables telecommunications service providers worldwide to build and manage customized, flexible and scalable IP telephony rollouts. From VoIP Managed Services featuring best-of-breed VoIP systems to its award-winning Cortex® middleware, VoIP Logic provides a comprehensive set of ROI-sensitive solutions for service providers looking to use VoIP technology.

## **VoIPConsultants.biz, LLC (343)**

**P / F: 847-230-9225 / 847-352-5351**

**[www.voipconsultants.biz](http://www.voipconsultants.biz)**

VoIPConsultants.biz, LLC is the leading agency for private label VoIP services in the United States. Our founder has 4+ years experience in private label VoIP and can help you understand the requirements of getting into the business. So whether its a turnkey solution, one that integrates existing carriers, or you require customized development, we have you covered.

## **VoIPstreet (243)**

**P: 1-888-204-VOIP (8647)**

**[www.VoIPstreet.com](http://www.VoIPstreet.com)**

Your total VoIP solution. Visit us at Booth #243 to learn more.

## **Volt Delta Hosted Solutions (306)**

**P: 877-899-9950**

**[www.voltdelta.com](http://www.voltdelta.com)**

VoltDelta Hosted Solutions brings together VoltDelta's 30 years of experience in delivering Directory Assistance to all of the major telecom companies in North American and Europe with our state of the art contact center platform, applications and functions.

At Contact Center 2.0 VoltDelta Hosted Solutions is announcing OASIS On-Call: hosted range integrated products and platform for automated speech and contact center services, designed to enable organizations to achieve rapid implementation, take up and usage via a per seat and or minute model without high capital investment. Some of the products in the On-Call portfolio include:

OASIS Voice Self-Service is a next generation IVR solution including CTI func-

tionality, Speech Recognition + TTS options, and hosted voice applications built to reduce agent call time and increase caller satisfaction.

OASIS Web Station comprises solutions that enable organizations to extend agent functionality regardless of where they are located and integrate into existing software and systems.

## **WBS Connect (543)**

**P / F: 866-927-2669 / 720-528-7849**

**[www.wbsconnect.com](http://www.wbsconnect.com)**

Since 2002, WBS Connect has been providing high bandwidth Internet connections and data networking services worldwide. WBS Connect leverages expert network engineers and global carrier partnerships to customize flexible and scalable customer solutions. Product and service areas include:

- IPv4 and IPv6 Gig and 10Gig Internet access
- Wide area network transport
- Worldwide data center colocation
- Custom network design, build, and management consulting

With its corporate headquarters in Denver, WBS Connect has an extensive technology framework built throughout North America, Europe, and Asia. The company is also working to expand its presence in South America and the Middle East. WBS Connect continues to deliver premier solutions to customers, making its mark in the industry with consistent yearly revenue growth in the triple digits.

## **Wyde Voice, LLC (532)**

**P / F: 866-508-9020 / 562-437-1422**

**[www.wydevoice.com/index.html](http://www.wydevoice.com/index.html)**

WYDE develops 16KHz, wide-band voice applications for companies of all sizes. We offer turnkey products with customizable features at the most cost effective price point on the market. We specialize in bringing high-quality voice to complex, multi-speaker / multi-listener environments.

## **XO Communications, Inc. (635)**

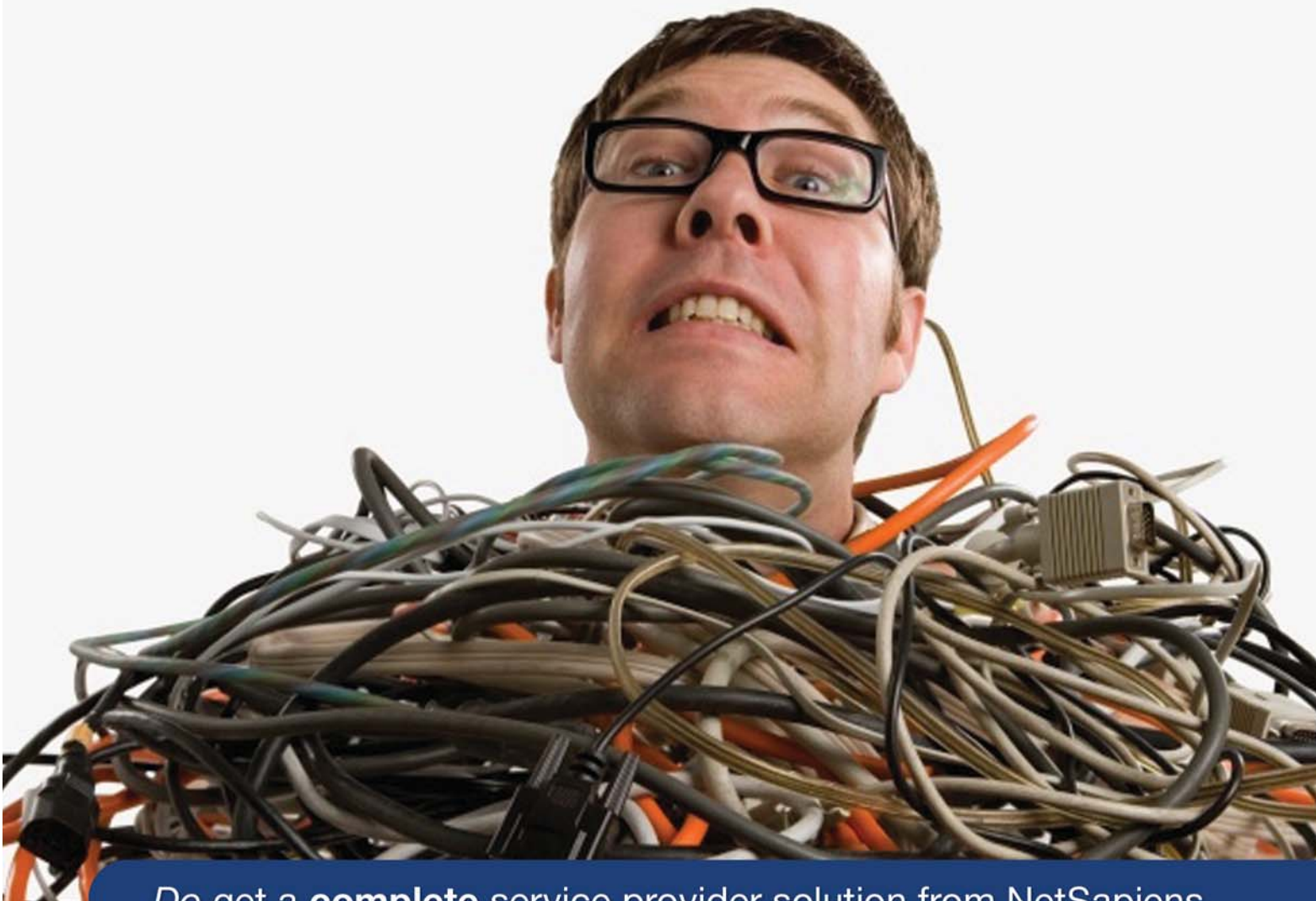
**P / F: 703-547-2000 / 703-547-3611**

**[www.xo.com](http://www.xo.com)**

XO Communications offers small, medium, enterprises and government customers advanced IP and networking solutions to maximize performance and simplify management of their networks. And, when it comes to implementing high-



*Don't get your wires crossed.*



Do get a **complete** service provider solution from NetSapiens.

**NETSAPIENS®**

**WE MAKE VOIP A SNAP™**

**Come see us at Booth #425**

3914 Murphy Canyon Road, Suite A100, San Diego, CA 92123 · (+1) 858-764-5200 · [www.netsapiens.com](http://www.netsapiens.com)



(Booth Number in Parentheses)

(Communications Developer Exhibitors highlighted in Green)

bandwidth applications or migrating to a converged network, XO leverages its extensive nationwide and metro networks to deliver end-to-end solutions and high-demand enterprise services.

## **Xorcom (OS 04)**

**P / F: 866-XORCOM-1 / 623-516-8697**

**[www.xorcom.com](http://www.xorcom.com)**

Founded in 2004, Xorcom is a privately-held IP-PBX manufacturer. Xorcom harnesses the power of Asterisk® Open Source IP-PBX – the most rapidly developing telephony platform in the world – to design and produce leading-edge hardware telephony solutions for commercial installations. Today, Xorcom offers the widest and most flexible range of solutions and hardware platforms in the Asterisk market. System integrators, telecom equipment manufacturers, and customer premise telephony and VoIP providers use

Xorcom products to provide added value to their end users.

Xorcom engineers are part of the worldwide Asterisk development team and cooperate with the main players in this field. The Astribank driver (xpp-Xorcom Peripheral Protocol) has been a standard component in the DAHDI (formerly Zaptel) package since the release of Asterisk version 1.2.4 in February 2006.

Xorcom sells its products via a worldwide distribution channel, OEM partners, and a sales office in the USA.

## **Zeacom Communications Center (520)**

**P / F: 800-513-9002 / 800-513-5874**

**[www.zeacom.com](http://www.zeacom.com)**

Zeacom has successfully pioneered award-winning software applications that deliver Unified Communications solutions for organizations with up to 2500 desktops, and contact centers with between 5

and 500 agents. More than 2500 SMBs across 25 countries use Zeacom's business communications software to improve communication, collaboration and productivity. For more information about lead products such as rich and mobility-enabled presence, or SIP-based conferencing, visit [www.zeacom.com](http://www.zeacom.com)

## **Zehu Technologies (247)**

**P / F: 866-938-9348 / 1-413-641-3430**

**[www.zehu.com](http://www.zehu.com)**

Zehu develops biometric voice verification technology that enhances security far beyond traditional password authentication. We pride ourselves on providing our OEM partners with the tools they require to easily integrate the Zehu technology into their products. If you are looking to enhance your current application or platform with the latest in authentication technology, come visit us and learn more.

## **Visit the Exhibit Hall.**

### **Find Solutions and You Could Win a New Prius Hybrid!**

1. Pick up your entry card at the registration counter when you arrive.
2. Visit each booth listed on the card, review their offerings, and receive a stamp from each booth.
3. Collect all stamps, then drop your completed card into the entry bin in the exhibit hall.
4. **\*YOU MUST BE PRESENT AT THE DRAWING ON THURSDAY, SEPTEMBER 18th. 4:55 PM TO CLAIM YOUR PRIZE!**

### **Prius Giveaway Sponsors:**



**\*2008 Toyota Prius Hybrid Drawing -  
Thursday, September 18, 4:55pm, at the  
conclusion of Exhibit Hall hours.**





**TelCentris™** ' award winning **Unified Communications Engine** enables the launch of a **profitable** IP Telephony business in a matter of days (*and it won't break the bank*)...

Applications:

- Carrier Grade Hosted Softswitch
- Full Class 4 and Class 5 capabilities
- Flexible Least Cost Routing
- Unified Messaging
- Business & Residential features
- Mobility features
- Calling Card, SMS & Web Callback
- Instant Messaging & Text Messaging
- Customizable Customer Facing Web Portals
- Web Based Administration
- Turn-key setup
- Pre & Postpaid billing
- Partitionable (create white-label resellers)
- Carrier Services available (L3, X0, Global Crossing, and more)
- DIDs/Termination/Toll Free/LNP/E911/E411/more

Please visit us at **booth #343**

Telcentris™ Inc. [www.telcentris.com](http://www.telcentris.com)  
1-866-612-8647 / 858-952-0692 / [clientdev@telcentris.com](mailto:clientdev@telcentris.com)



**telcentris™**  
SETTING COMMUNICATION FREE





## A South Carolina Corporation

NETXUSA has two distribution and sales centers located in the United States. East and West coast facilities provide rapid regional response and economical product delivery to the dealer channel.

NETXUSA provides cost-effective provisioning & branding services for our wide array of hardware direct from leading VoIP manufacturers. By only selecting proven VoIP hardware products, such as phones, analog adapters and gateways from our manufacturers, you can be confident that your application will be successful. NETXUSA backs all of our products with advanced replacement warranties and support services.

NETXUSA in-house certified engineers and an experienced customer service staff are available to service the dealer after the sale.

If IP enabling an existing legacy telephone system, planning a new VoIP PBX, or managing your Hosted VoIP Network is your challenge, NETXUSA has a solution for you.

For more information, visit us at booth 206.

To request reseller status, visit our New Reseller Sign-up at [www.netxusa.com](http://www.netxusa.com) or contact [sales@netxusa.com](mailto:sales@netxusa.com).

NETXUSA • PO Box 3599 • Greenville, SC 29608 • 800-289-6389



Reseller's choice for the leading VoIP products.

Why?

- \* We sell to Resellers only; End-Users are referred to the Reseller channel
- \* Recognized as the leading distributor of Digium and Asterisk related products
- \* Certified engineers on all products sold
- \* East & West Coast distribution centers
- \* Knowledgeable sales staff
- \* Blind drop shipping available
- \* Free tier II and tier III support
- \* Warranty handling with advanced replacement coverage on all purchases



- \* Complete IP PBX packages available



- \* Certified Asterisk & VoIP training center in Las Vegas, NV



- \* Phone, ATA, and router provisioning solutions







Aculab's Chris Gravett Discusses Company, Communications Market

TMC president and ITEXPO Conference Chairman Rich Tehrani recently had

the opportunity to ask Chris Gravett, Sales and Marketing Director at Aculab about the company's achievements and goals over the next year as well as what Aculab will be exhibiting at ITEXPO this week.

Aculab offers a portfolio of enabling technologies for VoIP developers and systems integrators to help them produce communications solutions that bring value, reduced costs, increased customer satisfaction and a competitive advantage.

**RT:** What has been your company's biggest achievement in 2008 so far?

**CG:** Our biggest achievement has been the success of Prosody X, established as the 'technology of choice' in four major sectors; conferencing, fax, contact center and military communications. Prosody X is now clearly the global market leader for communications enabling technology used in IP, hybrid and TDM converged communications solutions. Aculab's record breaking performance in fiscal year 07/08 (12 months ending June 2008) reflects the outstanding success of Prosody X.

The launch of the ApplianX range of products has also been extremely well received. The ApplianX IP Gateway and ApplianX Gateway for Microsoft Office Communications Server 2007 have both created much interest and demand in their respective target markets.

**RT:** What can we expect to see from your company for the next 12 months?

**CG:** We focus on our customers and we are willing to listen and invest in the technology they need to be successful. One of the key areas that sets Aculab apart is our ability to respond to customers' near-term requirements through our Request for Change (RFC) process. This process is designed to allow our customers to ask for product enhancements to target specific opportunities and is probably the key point that is brought out when our customers refer to Aculab as providing good support.

In terms of new products and features on the horizon, here are a few things that you can

look forward to seeing from Aculab in the coming year:

- Video support on our Prosody media processing family
- Improvements to licence management for Prosody S, our HMP product
- T.38 fax support
- Prosody S with Linux support
- Further variants within the ApplianX range

**RT:** How do you see the communications market evolving?

**CG:** The communications business model is now undoubtedly IP centric. Carriers have moved to IP in their core networks and, in spite of doubts over the relative penetration of IP into the user base, over 70 percent of equipment lines shipped support IP. And, despite what the H.323 Forum might say, I'm convinced the majority of these are SIP.

Mobile penetration has increased, too. In Singapore, for example, it's 125 percent and elsewhere in the world, it's also over 100 percent as many users have more than one SIM. Worldwide penetration now exceeds 50 percent. At well over 3 Billion, the number of mobile phones on the planet equates to over half the global population number.

More than 3 out of every 10 Americans now rely solely on a mobile handset for their telephony needs. Significantly, more than 84 percent of Americans are apparently now wireless subscribers. And, with 2007 becoming the year when worldwide mobile shipments exceeded 1 Billion for the first time, I predict an entirely mobile future.

Further evidence for this comes from mobile carriers as they determine how to move from GSM to LTE and/or WiMAX for their 3.5/4G services. Voice usage is shifting and it won't be long before more than two-thirds of traffic is wireless. It's not hard to deduce that there is a market for mobility and there is no question we are in a rising industry that is going to be increasingly powered by IP, broadband and some kind of mobile device.

**RT:** What will attendees see in your booth on the exhibit hall floor?

**CG:** We will display our flagship products, the Prosody X family of media processing boards. For developers of VoIP and PSTN telephony solutions, Aculab's Prosody product range provides feature rich and high-density media processing, call control signaling and digital network connectivity.

We will also display GroomerII, a system level performance packed product, based on our

Prosody X boards. GroomerII is primarily a signaling and media gateway for carriers and service providers, ideally suited to SS7-to-SIP interconnection.

The attendees will see the display of ApplianX IP Gateway and ApplianX Gateway for Microsoft Office Communications Server 2007. ApplianX is a range of deployment-ready appliances designed for infrastructure roles in converged telecommunications networks of enterprises, data centers and service providers.

**RT:** What sorts of companies should come to your exhibit?

**CG:** Anyone who is developing multimedia applications and solutions for IP, PSTN and mobile networks should visit the Aculab booth. We have the complete range of products and services to meet your needs. Aculab has the most comprehensive technology suite to satisfy companies who are servicing a host of sectors including: call progress analysis; contact centers; fax broadcast; conferencing; MAG and emergency services. These examples reflect some of the many applications and services where Aculab's communications products and core competencies have delivered significant added value to our customers and partners.

**RT:** Why should customers choose your company's solutions?

**CG:** Aculab not only has the most comprehensive suite of IP and TDM media processing products to satisfy our customer needs, but also offers the broadest range of pre- and post-sales support services. Key differentiators include: bespoke pre- and post-sales professional services, consultancy and training; unlimited post sales technical support, a Web-based knowledge base and interactive user forum; and a long established partner marketing program, ensuring that our customers achieve success at every stage of their product's life cycle.

In addition, Aculab is proud of its longevity—celebrating 30 years in business this year—private ownership and independent status, which all mean true flexibility can be exercised to meet our customer's needs at every opportunity.

Aculab is a Diamond sponsor of Internet Telephony Conference & EXPO — the biggest and most comprehensive IP communications event of the year. Visit Aculab at booth #314 in the exhibit hall.



# You've made the decision to private label VoIP services to your customers.



**The next call you make could be the difference between success and failure.**

VoIPConsultants.biz, LLC is the leading agency for private label VoIP services in the United States. Our founder has over ten years experience with VoIP products and services and 4+ years in private label VoIP. We offer more choices and services to meet our clients needs and we work hard to structure agreements that our clients and

carriers can live with. So whether its a turnkey solution, one that integrates existing carriers and applications, or you require customized development, we have you covered. If you need reliable, high quality service using state of the art technology to run your business efficiently and profitably, you've come to the right place.

**Visit Us At Booth 343**



P.O. Box 68934 • Schaumburg, IL 60168  
www.voipconsultants.biz • Tel: 847-230-9225





## Discover ABP's IP Solutions for Technology Partners.

ABP Technology is the leading Distributor for Open Standards Based IP Communication solutions specializing on IP Telephony, IP Surveillance and IP Peripherals. ABP sells and supports select products and solutions through a National Network of Specialized Resellers.

**Visit ABP Technology at Booth #618 for more details.**





## ABP Solution Packages Provided by ABP Technology Reseller Partners Nationwide:



### Business Phone System

Different types of businesses have different needs and Technology Partners add a lot of value in helping customers choose the right solution. We have made it easier by pre-selecting and testing the best and pre-engineering solution packages that contain many unique selling points for specific target groups.

ABP Technology has created special Telephony Packages to help you focus specific customer profiles with targeted pre-engineered products packages and marketing material offering the unique selling points for the target customer as well as all most frequently needed options to help you up-sell. Ask for our Telephony package overview.



### IP Surveillance

IP Surveillance solutions integrated into IT infrastructure, Self Monitoring or External Monitoring, Automatic Alarm responses including calls to cell phones or pagers with pre-recorded messages.

- State of the art Monitoring software
- IP Access Control Solutions

- Face Recognition Application
- Situation Management Center Software

- License Plate Recognition & Recording

ABP's IP Surveillance packages include:

- Small Open Spaces
- Open Lot
- Large Open Spaces
- Small Office



### IP Paging & Public Addressing

Create a universal paging environment:

- Integrate different platforms
- Deploy new IP Paging on legacy PBX

- Connect legacy paging to IP PBX
- Paging Zones

- Interconnect with Alarm systems
- Integrate with IP Video Surveillance



### Redundant Internet

Resellers become guardians of comparing Internet connectivity. ABP Redundancy solution will prevent Internet failure, lost productivity and will automatically switch packets over surviving service/router and contact reseller and customer in event of failure of a circuit.

- Multi honed Internet Service Based
- Wireless Backup Network Solutions



### Business Continuity Solutions

Hosted back-up phone solution that automatically kicks in when connectivity to office is lost or phone lines don't answer.

- Wireless Backup lines
- Backup PBX with routing to office over redundant data paths, routing to cell phones and backup voicemail and unified messaging



### Custom Telephony

IP PBX with integration to ERP or CRM:

- Call Center Applications
- Custom IVR
- Skype in and softphone solutions

- Unified Messaging
- Conditional look-ups and text to Speech
- Least cost routing over SIP Trunks and Wireless and cell plan integration

- Click to Call
- Call recording



### Mobility Solution for Office

- Wi-Fi Phones for Office and Home
- DUAL Wi-Fi Cell IP Phone Solutions

- DECT Cordless IP Phones
- Comprehensive Cell Phone Integration including LCR for in Plan Cell phones to IP PBX with GSM gateway



### Wi-Fi Solution - Connectivity for Campus, Resort, Metro & Business

- Wi-Fi Access for General Public or Business Only.
- Use ABP's unique ALTAI-A8 Base Station to build a secure and robust outdoor Wi-Fi Network with unprecedented range.
- Resort and Hospitality customers can use this as a revenue generator. QoS for wireless IP Surveillance camera and VoIP Phone solution.



### The Teleworker

One of the fastest growing needs is solutions for employees working from their home office. ABP has prepared the perfect Teleworker solution for SMB and Enterprise.

- QoS on ADSL for home workers
- 911 Filtering or E911 solution for Teleworkers

- High Definition Speaker Phone
- DECT Cordless IP Phones

- Comprehensive Cell Phone Integration



### The Sales Team Enabler Options

Companies that have travelling sales people are looking for ways to improve communications amongst the office and the team while reducing cost. ABP has a complete set of solutions including Unified Communications and a Skype interface enabling laptops to communicate through firewalls with the company's IP-PBX solution.

This same SKYPE gateway is also being used a lot by companies that export and have many international customers that want to call in via SKYPE.





**MyGlobalTalk™**  
connect with the world

***NO WI-FI or HOT SPOTS NO CONNECTION FEES***  
***NO MINIMUMS NO MONTHLY FEES***

***CALL ANYWHERE IN THE WORLD***  
***for as low as 2 cents per minute from any phone!***

i2Telecom® provides high-quality international and domestic long distance calling services to subscribers at a fraction of the cost of traditional carriers by leveraging the power of the Internet.

**[www.MyGlobalTalk.com](http://www.MyGlobalTalk.com) | [www.i2telecom.com](http://www.i2telecom.com)**





## ***Award Winning Technology***

*See us at Meeting Room #2*



***MyGlobalTalk* 2008 Best of Show-Internet Telephony Conference and Expo**



***MyGlobalTalk* 2007 Product of the Year-Internet Telephony Magazine**



***MyGlobalTalk* 2007 Unified Communications-Product of the Year**



***MyGlobalTalk* 2007 Communications Solutions-Product of the Year**



***i2Telecom*® 2005 Internet Telephony Magazine VoIP Service Provider of the Year**



***VoiceStick*® 2004 Internet Telephony Magazine award for Product of the Year**

[www.MyGlobalTalk.com](http://www.MyGlobalTalk.com) | [i2Telecom.com](http://i2Telecom.com)

© 2008 i2Telecom International, Inc.





COMMETREX ... *Open Service Platforms & Technologies For the New Network*

*What do all these companies have in common?*

2Wire, 3Com, 8x8, Altigen Communications,  
Amazon.com, AudioCodes, Avaya,  
Broadsoft, Basis-AudioNet, Castelle, Ciena,  
Copia International, CTI<sup>2</sup>, Cycos AG, Data  
Connection, Datera, Dialogic, Digium, E-Telko,  
emFAST, Empirix, eServGlobal, Freescale,  
Genband, General Dynamics, Grandstream,  
HelloSoft, InfiNet Wireless, iNetWorx AG,  
Inmarsat, Interactive Intelligence, Kapsch AG,  
Logica, LSI, Marvell, Motorola, Movius,  
Netergy, NetHawk, Nortel, Octasic,  
Openwave, PMC-Sierra, Pika Technologies,  
RadiSys, RingCentral, Sagem-Interstar,  
Shoretel, Siemens, SiTel Semiconductors,  
Sonus, Spirent Communications, Streem  
Communications, Tektronix, Teles AG, Trinity  
Convergence, Unicoi Systems, UTStarcom,  
Zhone Technologies





*They all have products  
Powered by Commetrex  
technologies*

For more information about Commetrex **BladeWare™** fax and voice applications, and media technologies, visit us at ITEXPO West (booth #127) in Los Angeles, September 16 - 18, 2008. Contact us at 770-407- 6020 or [sales@commetrex.com](mailto:sales@commetrex.com)



# TDM-to-SIP Gateways. To Go.



## applianx

It really **is** that easy. The ApplianX IP Gateway and ApplianX Gateway for Microsoft Office Communications Server 2007 offer an unbeatable combination of simplicity and resilience. Utilizing acclaimed enabling technology from Aculab, ApplianX gateways are the perfect means to execute your VoIP migration strategy.

**Configure your ApplianX in the time it takes to deliver a pizza.**

[www.applianx.com](http://www.applianx.com) | [info@aculab.com](mailto:info@aculab.com) | Booth 314



**Microsoft**  
GOLD CERTIFIED  
Partner