

SWITCHVOX FAMILY

Communication is the key to keeping your business running smoothly, and with a Switchvox™ system you can count on having the flexibility you need at a price you can



Switchvox SMB Redundant Rackmount System

afford. Switchvox is a complete business phone system in software. This means you can say goodbye to the monolithic PBX of the past. Now you can have the freedom to grow from a single office to a global network without breaking the bank. And because Switchvox systems are compatible with today's high speed data networks and traditional telephony infrastructure, you can change at your own pace.

There is an award winning Switchvox system that can meet your needs. Switchvox Free Edition (that's right, FREE!) can be downloaded and setup quickly. Free Edition gives you the chance to see what a Switchvox system can do for you. After you fall in love with your Switchvox system, you can easily upgrade to Switchvox SOHO or Switchvox SMB.

Switchvox SOHO comes with an easy web interface, unlimited extensions, and a host of features not seen in the Free Edition. Switchvox SOHO delivers an incredible set of enterprise class features to even the smallest offices, even when your're out of the office. Switchvox SMB is a complete communication system that can adapt to your dynamic enterprise. Switchvox SMB includes all the features of SOHO, as well as call recording, conferencing, and call queues.

If an investment in hardware is not in your budget, or you are unsure how quickly your needs may grow, Switchvox Hosted Edition may be for you. Switchvox Hosted Edition can be inexpensively deployed in a data center on a high-density server. Once you are ready for an on premise solution, such as with Switchvox SOHO or SMB, your configuration can be exported seamlessly.

With the Switchvox family of products you've got an easy to setup PBX solution, no matter the size of your company or how much you grow!

"C" OUT NEW EDITION

Asterisk Business Edition™ is the enterprise-grade version of Asterisk®, our acclaimed open source telephony tool-kit. Release C provides a tested and fully supported telephony applications platform based on the 1.4 release of Asterisk and incorporating musthave features from the current development branch. Bundled with a zero-effort Linux™ installation from rPath®, the Asterisk Graphical User Interface (AsteriskGUI™) and a one year support subscription, Business Edition Release C makes voice system development a snap. The easy-to-use GUI drastically simplifies the process of administration. Hundreds of improvements to the code and thousands of hours of testing make ABE a solid business decision.

Version C also includes the following new features:

- Upgraded LumenVox® and Cepstral Support
- Digital Card Detection and Configuration (GUI)
- Increased Dialplan Functionality
- Easy-to-use Graphical Administration System
- Polycom® Phone Provisioning

In addition to rPath™ Linux™ v1 support,
Asterisk Business Edition C is supported on:
Fedora™ Core 6 (32 and 64 bit), Fedora 7 (32 and 64 bit), RHEL4 ES Update 4 (32 and 64 bit), RHEL5 ES (32 and 64 bit). A new manual for Asterisk Business Edition C will also be available. This manual includes contributions from Digium as well as the authors of Asterisk: The Future of Telephony!



Version C is the next major step in the evolution of the enterprise-grade software which has revolutionized open source communication. For more information on Asterisk Business Edition C, please visit http://www.digium.com/en/products/software/abe.php

NEW PRODUCTS

AEX2400 //

An express analog gateway

The Digium® AEX2400 is the PCI-Express compatible version of our popular TDM2400 card. The AEX2400 is a full-length PCI-Express 1.0-compliant modular gateway card for connecting analog telephone stations and analog POTS lines through a PC. It supports a combination of up to six quad port station or trunk modules for a total of 24 lines. The AEX2400 eliminates the requirement for separate channel bank and T1 interface cards, with industry-leading performance and price. An optional hardware echo cancellation module, Digium's VPMADT032, provides 1024 taps (128 milliseconds) of echo cancellation for superior voice quality on both trunk and station interfaces.



TE121 & TE122 // Select a new single span communications channel

The TE121 and TE122 are single span, selectable T1 (24-channel), E1 (32-channel), or J1 (24-channel) cards. The TE121 is a PCI-Express 1.0-compatible version of the TE120p. The TE122 is an update to the previously released TE120p. Both cards improve upon the TE120p by offering expansion capabilities. The cards can be combined with Digium's VPMADT032 echo cancellation module to deliver 128ms of echo cancellation across 30 channels in E1 mode or 24 channels in T1 mode.



Additional information and pricing can be found at: www.digium.com/en/products/hardware



fundraising, Mark Spencer had to kiss a real pig, and was not too thrilled about it.

Check out the video at www.makeadigiumwish.com

Digium[®] | Asterisk[®] Events

Digium|Asterisk World™ @ Spring VON

March 17-20, 2008 San Jose, California — USA

Digium | Asterisk World is the premier open communication event for the channel that encompasses the world of open source platforms and applications in the realm of IP communications. Whether you are a service provider, VAR, systems integrator or someone who is rolling out IP communications internally, Digium | Asterisk World is the place to be.

www.digiumasteriskworld.com

To exhibit, contact Christopher Erb at christopher.erb@pulvermedia.com

VoiceCon Orlando

March 17-20, 2008 Orlando, Florida — USA

Digium|Asterisk World™ @ VON Europe

June 2-5, 2008 Amsterdam, Netherlands — Europe

NXTcomm

June 16-19, 2008 Las Vegas, Nevada — USA

Astricon

September 22-25, 2008 Phoenix, Arizona — USA

Digium|Asterisk World™ @ Fall VON

October 27-30, 2008 Boston, Massachusettes — USA

VoiceCon San Francisco

November 10-13, 2008 San Francisco, California — USA

AstriCon is the official conference for the wildly popular Asterisk® PBX and telephony platform. AstriCon includes a wealth of information for those who are getting started with Asterisk, as well as those who have already discovered the power of open source telephony.

For 2008, Astricon remains in Phoenix, Arizona, but will be located in a high-rise hotel instead of spread out across a large resort. The Renaissance Glendale Hotel and Spa will be home to the 2008 Astricon Conference September 22nd - 25th.

www.astricon.net

BIGIUM WINS

InfoWorld recognizes Asterisk® 1.4 as the best IP PBX in 2008



Selected by InfoWorld Test Center editors and reviewers, the annual awards, presented this year on January 8, identify the best and most innovative products on the IT landscape. Asterisk® is the most widely used open source telephony software with more than 1 million downloads. in 2007 and more than 100 million users worldwide.

InfoWorld reports, "a complete IP PBX released as open source under the GNU Public License, Asterisk is built to run on commodity hardware, providing considerable cost savings when compared with commercial alternatives, and it leverages the open source community for additional testing, bug fixes, and feature development." For a full list of award recipients, see www.infoworld.com.

"InfoWorld tests upwards of 200 IT products every year, and we see many, many good ones," said Doug Dineley, InfoWorld's Test Center executive editor. "Our Technology of the Year award winners represent not only the cream of the crop, but the best products in the most important product categories. From the top AJAX development tools and SOA middleware to the best blade servers and VoIP systems, these are the products at the leading edge of IT."

Mark Spencer, creator of Asterisk and Digium's chief technology officer, commented: "The InfoWorld Technology of the Year 2008 award is an enormous acknowledgement of the work and success of the worldwide Asterisk community. Asterisk wouldn't be in a position to rival some of the biggest vendors of proprietary voice technology today if it weren't for the inherent strength of open source and the dedication of the community developers."

On December 19th, 2007 Digium® announced the

ONE-MILLIONTH DOWNLOAD OF ASTERISK IN 2007.

2007's Q4 was also Digium's

24TH consecutive quarter

of growth & profitability.



ASTERISK



Asterisk® Bootcamp: A 5-day Ultra-Intensive Course

February 18 - 22, 2008 Huntsville, Alabama, USA, North America

February 18 - 22, 2008 Anaheim, California, USA, North America

February 18 - 22, 2008 Mount Waverly, Victoria, Australia

March 10 - 14, 2008 Huntsville, Alabama, USA, North America

March 10 - 14, 2008 Aylesbury, Buckinghamshire, UK, Europe

March 10 - 14, 2008 Venissieux, France, Europe

March 10 - 14, 2008 Las Vegas, Nevada, USA, North America

March 17 - 21, 2008 Paris, France, Europe

April 7 - 11, 2008Anaheim, California, USA, North America

April 14 - 18, 2008 Huntsville, Alabama, USA, North America

May 5 - 9, 2008 Huntsville, Alabama, USA, North America

Classes are periodically added throughout the year. Please check the Digium® website for a complete and up-to-date list.

www.digium.com/en/training



new ASTERISK® Training Program

Training program updates to build on already successful program.

Over the first quarter of 2008, Digium will deliver 15+ Bootcamp classes all over the world. Digium has consistently maintained a 4.5 Customer Service Rating with it's attendees, and will continue to strive for excellence. Our existing Bootcamp materials have been updated in order to be current with our technology.

Keeping in step with providing the best customer experience possible, we are in the process of establishing a new Learning Management System that will take into account localized registration procedures in collaboration with our 2008 Authorized Training Partners. For starters, from within the LMS, one will be able to select and participate in a CBT environment, access the latest Videos on Demand of our SwitchVox™ product line, and access entry level Asterisk® 101 courses. Moreover, we are creating a 2 Day-Mini Bootcamp that will focus on the needs of the financial decision makers from the SMB owners to the Enterprise leaders.

The Digium Authorized Training Partners will be able to access their training materials from the Digium Training Portal. One of our newest ATPs is David Duffett. David's company, TeleSpeak Limited, located in London, will be among the first localized Bootcamps to be delivered, focusing on the UK specific Asterisk requirements.

Continuous improvement of the existing dCAP Test Pool with new questions/answers from around the globe is a top priority. Authorship validation will be given to those who participate and have their questions selected by a Digium sponsored panel. Once this activity is completed, Digium will be able to host the dCAP on-line. Please submit your questions/answers in English to ssalins@digium.com.

Any training tools or successful training stories that can be shared are welcome and valued by all. The Digium Training Team looks forward to meeting you and will continue to offer the dCAP at selected events.

STOP SHOP

Digium® Unveils Business-Class Digium® | Asterisk® Marketplace

Digium® launched the "Digium® | Asterisk® Marketplace" on January 8th to give the Asterisk® community of end-users, developers, small businesses and enterprises a one-stop online destination for all their open source VoIP deployment needs. The Digium | Asterisk Marketplace enhances Digium's Technology Partners web portal to allow companies that are not yet Digium partners to advertise their products to the Asterisk Ecosystem.

Visitors to the Digium|Asterisk Marketplace will be able to search the company listings by keyword to easily find Asterisk-compatible products such as phones, network infrastructure devices, and application software, as well as services such as VoIP and PSTN providers. The website opens up a new, direct line of communication between independent vendors of Asterisk-related products and the entire Digium|Asterisk ecosystem.

"The Digium website generates thousands of hits per day as Asterisk adoption continues to grow in business

environments around the world," said

Jim Webster, director of technology

partnerships for Digium."We expect the Digium|Asterisk Marketplace to become a premier destination for companies evaluating and selecting offerings to help build complete Asterisk-based solutions. Digium is excited to be working with leading technology partners and vendors to make it easier than ever for businesses to create open source telephony solutions."

To learn more about the Digium|Asterisk Marketplace please visit www.digium.com/marketplace, or e-mail marketplace@digium.com.



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