Polycom[®] VSX[™] 3000 Features and Benefits



	Audio
G.722.1 Annex C (based on Polycom Siren™ 14) Polycom StereoSurround™	 14 kHz crystal clear, wideband audio, while only using 24-48 Kbps of bandwidth, depending on the data rate of the call Eliminates fatigue associated with straining to hear every word Near CD-quality sound for dynamic audio performance Superb clarity to hear the subtleties of every word Rated the best audio algorithm in the industry; outperforming MPEG4 in all tests Included with every VSX system For more detailed information, see www.polycom.com/siren14 2 channels of 14 kHz crystal clear audio Superb clarity to hear the subtleties of every word Easily decipher multiple, simultaneous conversations Distinguish which side of the room people are talking from, just like
Audio Error Concealment	 in a real meeting Reduced audio drop-out on busy IP or ISDN networks Smooth, consistent verbal communications without interruptions Corrects audio from any system seamlessly Activates automatically, only when needed
H.264 Video	Video Use less bandwidth to conduct video calls of equal quality compared to calls using H.263 standard Unparalleled video quality for lower line rate applications Remote office advantage since less bandwidth is needed Cost savings to any organization
Pro-Motion™ H.263	 Supported at data rates from 64Kbps to 768Kbps Full-screen, TV-like video quality with fluid, precise motion handling Greater image detail with 50/60 fields per second Most natural video communications experience possible
Video Error Concealment	 Reduced video drop-out on busy IP networks and the Internet Smooth, continuous video without interruptions Maintain active face-to-face contact Accomplish meeting objectives without worrying about the video integrity Activates automatically, only when needed
People+Content™, H.239	Polycom People+Content or standards based H.239 to insure interoperability Dual images allows the far end to see the content and the speaker at the same time
People+Content IP	 Using a light utility on your PC, simply connect to your V500 and PC content is shown to all conference participants This feature can be used by anyone in the conference, even remote voice-only participants!
Adjustable Bandwidth for Content	 Choose variable bandwidth allocated to People or Content Allows setting quality preference of a 90% / 10% split between People and Content or sharing bandwidth equally between People and Content (50% / 50%) User adjustable setting can be accessed both during a call and outside of a call



	 Now all sources of content (PC, 2nd camera, VCR, etc.) show up on the second screen/monitor –never displace the person speaking Supplement video conference with multiple types of content
	including pictures, documents, audio files, presentations
	Flexible bandwidth settings for adjusting people and content
	ser Interface
Calendar & Call Scheduler	 Quick and easy access to a monthly calendar with current date and time posting
	 Schedule individual or repeat meetings to dial automatically Create and save multipoint team meetings
Customizable Home Page	Display only the necessary buttons
-	 Unique design to meet your organization's needs
	 Simplified graphical interface for novice users
	 Easy to understand icon driven menu navigation decreases learning curve
	 Kiosk Mode allows you to create a look and feel that suits your
	application
	Leads to increased use of video
Oual Monitor Emulation	 Use Dual Monitor Emulation for the most efficient use of a single display
	 View near and far video windows at the same time
	 Alternate layout views supporting near end, far end and content
Jser Selectable Camera Icons	 Use icons that represent your industry
	 Intuitive icons match the input device
	Custom name gives the icon clear meaning
	Selection – 5 libraries to choose from
	Decreases learning curve
	orm Versatility
ntegration with Microsoft® Live Communications Server (LCS) via SIP	 Integrates directly with Microsoft collaboration infrastructure Registers and authenticates with Microsoft LCS 2005
John Marine ations derver (EGG) via on	VSX users can be added to Microsoft Communicator
	 Presence information sent to LCS indicating video buddies'
	availability
	You can use your Buddy list to launch calls from the VSX 3000
	user interface
Completely integrated, all in one system	 Includes 17" display perfect for full screen viewing of remote and Picture-in-Picture for near end video
	 Integrated stereo microphones and speakers eliminates the need
	for table top microphones and external speakers
N ('' BOB'	Supports Polycom StereoSurround
Also functions as PC Display	 Use as a high resolution (SXGA) display for a local PC connection, saving space within the office
	PC audio can be connected to the VSX 3000 so the built-in
	speakers or headphones can reproduce the PC audio
P (H.323, SIP or SCCP) or ISDN/IP Models	 Choose the right model for your specific network connection
	SCCP – integration with Cisco Call Manager (in 128 MB systems)
	Supports UPnP and NAT for automatic setup of conferences
	conducted through firewalls
ntornal Multipoint Coffuers	Wide range of IP QoS services to insure call quality and integrity
Internal Multipoint Software	 Optional upgrade easily done with a software key
memai wuliipoini Sollware	Conduct meetings beyond just point to point
memai wunipoini Sonware	 Conduct meetings beyond just point-to-point Increase productivity by including key people on the call

	 Supports up to 4 video and 1 voice call in a single conference
Advanced Encryption Standard (AES)	Communications are confidential and secure Authentication by the National Institutes of Standards means it's credible Puilt in no outro bordware or coffware in required.
H.460 NAT/Firewall Traversal	 Built-in, no extra hardware or software is required Standards compliance for traversing NATs and Firewalls Session border controller (V2IU) is only needed at the central location and all VSX systems with H.460 can communicate through the single V2IU Uses H.460.18 for signaling traversal and call establishment and H.460.19 for media traversal
E-Mail dialing format when dialing across IP security boundaries	 Easy and intuitive E-mail video dialing Operates with Polycom's V2IU™ Firewall Traversal Appliance Deployable to wide range of customers, suppliers and partners
	Interoperability
Part of Polycom's Unified Collaborative Communications	 Remote management through Global Management SystemTM MGC Click & ViewTM offers a variety of layout templates for a multi point call right from the system's handheld remote Extended conferencing with Polycom PathNavigatorTM Schedule, invite participants and manage conferences easily with Polycom Conference Suite Use the Polycom SE200 for management, scheduling and gatekeeping
Configurable MTU packet size	 Allows administrators to set the MTU packet size, based on needs of their network Optimizes packet size minimizing overhead and network congestion Default setting is 1260
Standards-based	 Solid, reliable platform Qualified by independent test labs Interoperates with other vendor's systems Easy to install even in a multi-vendor environment See technical specifications for complete list of standards
FIPS 140-2 Certified	 Secure FTP, Telnet and Web VSX and V-series systems are the most secure video conferencing solutions in the industry Now, all access to the VSX system can be secured including accessing the VSX system via web-browser as well as secure telnet and ftp access via SSH for secure system administration and application connectivity
Baseline Mode	 Set your system to baseline mode, H.261 and G.711, for maximum interoperability with legacy video conferencing systems Accessible via web, FTP or on-screen UI for easy access by the network administrator

• Can be turned on and off as needed

