



DATA SHEET

Polycom® VVX® 101 Business Media Phone

An affordable and reliable single line IP desk phone

The Polycom® VVX® 101 is a simple, yet reliable, one-line SIP phone, with a 10/100 Ethernet port, that delivers enterprise grade sound quality. The VVX 101 phone is a stylish, cost effective telephony solution, ideal for a home office or shared/common areas, such as lobbies, hallways and break rooms or anywhere needing simple and reliable connectivity.

Unsurpassed voice quality and clarity

The VVX 101 full duplex Type 1-compliant speakerphone with legendary Polycom® Acoustic Clarity™ and Polycom® Acoustic Fence™ technology that delivers excellent sound quality and enables noise- and echo-free conversations that are as natural as being there.

Simplicity and ease-of-use

The Polycom VVX 101 comes with a familiar, intuitive user interface with multi-language support that you can use without having to think about the “how to”.

Best-in-class deployment and administration

The VVX 101 phone is engineered to make installation, configuration, and upgrades as simple and efficient as possible. An enterprise-grade, Web- based configuration method allows administrators to easily provision and maintain a large number of phones throughout an organization. Additionally, the extensive administrative feature set included ensures that hosted service providers have the latest technologies to integrate the VVX 101 seamlessly into their existing telephony environment.

Broad and robust interoperability

The VVX 101 is designed for interoperability and is certified to interoperate with the industry's broadest range of PBX systems and SIP call control platforms.



Benefits

- Use the VVX 101 as a simple, one line phone for home office or shared/common-areas
- Make more efficient and productive calls with unparalleled voice clarity
- Deploy in hosted or premise-based environments
- Reduce deployment and maintenance costs—the Polycom Zero Touch Provisioning and web-based configuration tool makes the VVX 101 simple to deploy, easy to administer, upgrade, and maintain
- Leverage previous IT infrastructure investments—deploy VVX 101 business media phones on your existing network without needing to upgrade your call control platform

Polycom VVX 101 Specifications

User interface features

- 2.5 in Graphical LCD (132 x 64) resolution
- Voicemail support
- Reversible deskstand/wallmount
- Unicode UTF-8 character support. Multilingual user interface including Danish, Dutch, English (Canada/US/UK), French, German, Italian, Norwegian, Polish, Portuguese, Russian, Slovenian, Spanish, and Swedish

Feature keys

- 4 context-sensitive “soft” keys
- 2 line keys with bi-color (red/green) LED
- “Home” feature key
- 4-way navigation key cluster with center “Select” key
- 2 volume control keys
- Dedicated hold key
- Dedicated headset key
- Dedicated hands-free speakerphone key
- Dedicated microphone mute key

Audio features

- Polycom Acoustic Clarity™ technology providing full-duplex conversations, acoustic echo cancellation and background noise suppression
 - Type 1 compliant (IEEE 1329 full duplex)
- Frequency response – 150 Hz – 7 kHz for hands-free speakerphone and optional headset mode
- Codecs: G.711 (A-law and µ-law), G.729AB, G.722
- Individual volume settings with visual feedback for each audio path
- Voice activity detection
- Comfort noise generation
- DTMF tone generation (RFC 2833 and in-band)
- Low-delay audio packet transmission
- Adaptive jitter buffers
- Packet loss concealment

Headset and handset compatibility

- Dedicated RJ-9 headset port
- Hearing aid compatibility to ITU-T P.370 and TIA 504A standards
- Compliant with ADA Section 508 Subpart B 1194.23 (all)
- Hearing aid compatible (HAC) handset for magnetic coupling to hearing aids
- Compatible with commercially-available TTY adapter equipment

Call handling features¹

- 1 SIP identity (registration)
- Additional programmable line key
- Shared call/bridged line appearance
- Flexible line appearance (one or two line keys can be assigned for phone's registration)
- Distinctive incoming call treatment/call waiting
- Call timer and call waiting
- Call transfer, hold, divert (forward), pickup
- Called, calling, connected party information
- Local three-way audio conferencing
- One-touch speed dial, redial
- Remote missed call notification
- Do not disturb function
- Electronic hook switch capable
- Local configurable digit map/dial plan

Open application platform

- Supports Polycom Apps SDK and API for third-party business and personal applications²
- Bundled with Polycom UC Software:
 - Corporate Directory Access using LDAP
 - Visual Conference Management

Network and provisioning

- SIP Protocol Support
- SDP
- IETF SIP (RFC 3261 and companion RFCs)
- Single 10/100Base-TX port
- Manual or dynamic host configuration protocol (DHCP) network setup
- Time and date synchronization using SNTP
- FTP/FTPS/TFTP/HTTP/HTTPS server-based central provisioning for mass deployments
- Provisioning and call server redundancy supported
- QoS Support – IEEE 802.1p/Q tagging (VLAN), Layer 3 TOS and Diffserv/DSCP
- VLAN - CDP, DHCP VLAN discovery, LLDP-MED for VLAN discovery
- Network Address Translation (NAT)—support for static configuration and “Keep-Alive” SIP signaling
- RTCP and RTP support
- Event logging
- Syslog
- Hardware diagnostics
- Status and statistics reporting

- TCP
- UDP
- DNS-SRV
- IPv4 and IPv6

Security

- 802.1X Authentication and EAPOL
- Media encryption via SRTP
- Transport Layer Security (TLS)
- Encrypted configuration files
- Digest authentication
- Password login
- Support for URL syntax with password for boot server address
- HTTPS secure provisioning
- Support for signed software executables

Power

- Built-in auto sensing IEEE 802.3af Power over Ethernet (Class 1)
- External Universal AC Adapter (optional, 12V 6W DC)

Approvals

- FCC Part 15 (CFR 47) Class B
- ICES-003 Class B
- EN55022 Class B
- CISPR22 Class B
- VCCI Class B
- EN55024
- EN61000-3-2; EN61000-3-3
- NZ Telepermit
- UAE TRA
- Australia RCM
- ROHS compliant
- ICASA
- CITC
- ANATEL³
- Customs Union³
- KCC³
- TAA³
- CCC³

Safety

- UL 60950-1
- CE Mark
- CAN/CSA C22.2 No 60950-1
- EN 60950-1
- IEC 60950-1
- AS/NZS 60950-1

Operating conditions

- Temperature: 0 to 40°C (+32 to 104°F)
- Relative humidity: 5% to 95%, noncondensing

Storage temperature

-40 to +70°C (-40 to +160°F)

Polycom VVX 101 comes with:

- VVX 101 console
- Handset with handset cord
- Network (LAN) cable - CAT-5E
- Quick Start Guide

Size

- 6.5 x 6 x 7 in (17 x 15 x 18 cm)(W X H X D)

Part numbers

- 2200-40250-025 – VVX 101 WW PoE

Unit weight

- 1.8 lbs (0.8 kg)

Master carton quantity

- Ten (10)

Warranty

One (1) year

Learn more

To learn more about VVX, visit

<http://www.polycom.com/voice-conferencing-solutions/desktop-ip-phones.html>

¹ Most software-enabled features and capabilities must be supported by the server. Please contact your IP PBX/Softswitch vendor or service provider for a list of supported features.

² Available in future UC Software release

³ Planned compliance and localization

About Polycom

Polycom helps organizations unleash the power of human collaboration. More than 400,000 companies and institutions worldwide defy distance with video, voice and content solutions from Polycom. Polycom and its global partner ecosystem provide flexible collaboration solutions for any environment that deliver the best user experience and unmatched investment protection.

