



Product Name: Polycom SoundStation IP5000 IP Conference Phone Manufacturer: -Model Number: 2200-30900-025

Please Note: This product has been discontinued. For an alternative, please see the other Poly Products.

Polycom SoundStation IP5000 IP Conference Phone The Polycom SoundStation IP5000 provides a cost-effective, HD sound quality solution for both small conference rooms and executive offices. The IP5000 features Polycom HD Voice technology, broad SIP interoperability, and a modern design that fits into any office environment. Polycom IP5000 Key Features

- ï¿¹/₂ Polycom HD Voice
- ï¿1/2 Polycom Acoustic Clarity technology
- ï¿1/2 3 sensitive microphones, with 360° coverage
- ï¿1/2 7ft microphone pickup for conference rooms with up to 6 participants
- ï¿1/2 Advanced IP feature support

ï¿1/2 Robust interoperability – compatible with most SIP call platforms

With Polycom HD Voice technology, the SoundStation IP5000 conference phone boosts productivity and reduces listener fatigue by turning ordinary conference calls into crystal-clear, interactive conversations. As the IP5000 captures all the frequencies of the human voice, conference calls sound natural without jitter or interference.

Polycom SoundStation IP5000 Conference Phone Technical Specifications Power

ï¿1/2 IEEE 802.3af Power over Ethernet (built in)

ïزئ Optional external universal AC power supply kit: 100-240V, 0.4A, 48V/19W

Display

- ï¿1/2 Size (pixels): 248 x 68 (W x H)
- i¿1/2 White LED backlight with custom intensity control

Keypad

- ï¿1/2 Standard 12-key keypad
- ï¿1/2 Context-dependent soft keys: 4
- ïزئ On-hook/Off-hook, redial, mute, volume up/down
- ï¿1/2 5-way navigation Menu

Audio Features

- � Loudspeaker
- ï¿1/2 Frequency: 250 7,000 HZ
- i¿1/2 Volume: Adjustable to peak volume 84 dB at 1/2 meter distance
- ï¿1/2 Voice activity detection
- ï¿1/2 Comfort noise fill
- � DTMF tone generation / DTMF event RTP payload



- ï¿¹/₂ Low-delay audio packet transmission
- ï¿1/2 Adaptive jitter buffers
- ï¿1/2 Packet loss concealment
- ï¿1/2 Acoustic echo cancellation
- ï¿1/2 Background noise suppression
- � Supported Codecs
- ï¿1/2 G.711 (A-law and µ-law)
- � G.729a (Annex A, B)
- � G.722
- � iLBC

Call Handling Features

- ï¿1/2 Shared call / bridged line appearance
- ï¿¹/₂ Busy Lamp Field (BLF)
- ï¿1/2 Distinctive incoming call treatment / call waiting
- ï¿1∕₂ Call timer
- ï¿1/2 Call transfer, hold, divert (forward), pickup
- \ddot{i}_{2} Called, calling, connected party information
- ï¿1/2 Advanced Local three-way conferencing
- ï¿1/2 (conference, join, split, hold, resume)
- ï¿1/2 One-touch speed dial, redial
- � Call waiting
- ï¿1/2 Remote missed call notification
- ï¿1/2 Automatic off-hook call placement
- ï¿¹/₂ Do not disturb function

Other Features

- ï¿1/2 Local feature-rich GUI
- \ddot{i}_{2} Time and date display
- ï¿1/2 Corporate Directory Access (search, dial,
- ï¿1/2 save to local directory)
- ï¿¹/₂ Convenient volume adjustment keys
- \ddot{i}_{2} User-configurable contact directory and call
- ï¿¹/₂ history (missed, placed, and received)
- ï¿¹/₂ Customizable call progress tones
- ï¿1/2 Wav file support for call progress tones

i¿¹/₂ Unicode UTF-8 character support. Multilingual user interface encompassing Simplified Chinese, Danish, Dutch, English (Canada / US / UK), French, German, Italian, Japanese, Korean,

- ï¿1/2 Norwegian, Polish, Portuguese, Russian,
- ï¿1/2 Slovenian, Spanish, Swedish

Network and Provisioning

ï¿1/2 Ethernet 10/100 Base-T

- ï¿1/2 IP Address Configuration: DHCP and Static IP
- ï¿¹/₂ Time synchronization with SNTP server
- تزنائ FTP / TFTP / HTTP / HTTPS server-based central provisioning for mass deployments.
- ï¿1/2 Provisioning server redundancy supported.
- ï¿1/2 Web portal for individual unit configuration
- i¿ $\frac{1}{2}$ QoS Support -- IEEE 802.1p/Q tagging (VLAN), Layer 3 TOS and DSCP
- ï¿¹/₂ Network Address Translation (NAT)
- ï¿1/2 support static



- ï¿¹/₂ RTCP support (RFC 1889)
- ï¿1/2 Event logging
- ï¿1/2 Local dial plan
- ï¿1/2 Hardware diagnostics
- \ddot{i}_{2} Status and statistics
- \ddot{i}_{2} User selectable ringer tones
- ï¿1/2 Field upgradeable

Security

- ï¿1/2 Transport Layer Security (TLS)
- \ddot{i}_{2}_{2} Encrypted configuration files
- ï¿1/2 Digest authentication
- ï¿1∕2 Password login
- $\ddot{\imath}_{\dot{\epsilon}} \overset{\prime \prime }{}_{2} \$ Support for URL syntax with password for boot server
- ï¿1/2 HTTPS secure provisioning
- ï¿1/2 Support for signed software executables

Safety

- � CE Mark
- � EN60950-1
- � IEC60950-1
- � UL60950-1
- ï¿1/2 CAN/CSA C22.2 No.60950-1-03
- � AS/NZS60950-1
- ï¿1/2 RoHS Compliant

EMC

 i¿½
 FCC Part 15 (CFR 47) Class B

 i¿½
 ICES-003 Class B

 i¿½
 EN55022 Class B

 i¿½
 CISPR22 Class B

 i¿½
 AS/NZS CISPR22 Class B

 i¿½
 VCCI Class B

 i¿½
 EN22024

Protocol Support IETF SIP (RFC 3261 and companion RFCs) Product Ships With

- ï¿1/2 Conference Phone Console
- ï¿¹/₂ 25 foot Ethernet cable
- ï¿1/2 Quick Start Guide

Optional AC Power Kit Ships With

- ï¿¹/₂ Universal Power Supply
- $\ddot{\iota}_{2}^{1/2}$ 7 foot region-specific power cord
- ï¿1/2 Power Insertion Cable

Environmental Conditions

ï¿1/2 Operating temperature: 41 – 104 degrees F



- ï¿1/2 (5 40 degrees C)
- i¿1/2 Relative humidity: 20% 85% (noncondensing)
- ï¿1/2 Storage temperature: -22 131 degrees F
- ï¿1/2 (-30 55 degrees C)

Warranty 1 year

Please Enquire

Options available for Polycom SoundStation IP5000 IP Conference Phone :

Power Supply Required?

IP5000 PSU (+87.31EUR), - Not Required -.