

Product Name: Polycom SoundPoint IP 560 VoIP Phone (IP560)

Manufacturer: Polycom Model Number: IP 560

Polycom SoundPoint IP 560 VoIP Phone

The Polycom SoundPoint IP 560 is a 4-line, desktop VoIP phone that delivers calls of unprecedented richness, clarity and supports a comprehensive range of cutting-edge features. SoundPoint IP560 Key Features

- Four SIP lines
- Backlit 320 x 160-pixel graphical grayscale LCD
- Shared call/bridged line appearance
- · Busy lamp field (BLF)– Presence, buddy lists
- XHTML micro-browser for productivity-boosting

Polycom SoundPoint IP 560 VoIP Phone - Technical Specifications Lines

• Up to 4 lines with up to 24 concurrent calls per line

#### Display

 320 x 160-pixel backlit grayscale graphical LCD LED backlight with custom intensity control Message Waiting Indicator (MWI) LED

#### Feature Keys

- 4 context-sensitive "soft" keys
- 26 dedicated "hard" keys 4 line keys with bi-color (red/green) LED 8 feature keys 6 display/menu navigation keys 2 volume control keys Illuminated mute key Illuminated headset key Illuminated hands-free speakerphone key Dedicated hold key

#### Headset Compatibility

Dedicated RJ-9 headset port - Amplified headsets are recommended

### Hearing Aid Compatibility

- Compliant with ADA Section 508 Recommendations: Subpart B 1194.23 (all)
- Hearing Aid Compatible (HAC) handset for magnetic coupling to approved HAC hearing aids
- Compatible with commercially-available TTY adapter equipment

#### **Audio Features**

- Polycom HD Voice technology delivers life-like voice quality for each audio path the handset, the hands-free speakerphone, and the optional headset
- Full-duplex hands-free speakerphone Type 1 compliant with IEEE 1329 full duplex standards
- Frequency response 100Hz 7kHz for handset, optional headset and handsfree speakerphone modes
- Codecs: G.722 (wideband), G.711 µ/A, G.729A (Annex B), iLBC
- Individual volume settings with visual feedback for each audio path
- · Voice activity detection
- · Comfort noise fill
- DTMF tone generation / DTMF event RTP payload

- · Low-delay audio packet transmission
- · Adaptive jitter buffers
- Packet loss concealment
- · Acoustic echo cancellation
- · Background noise suppression

#### Call Handling Features

- Shared call / bridged line appearance
- Flexible line appearance (one or more line keys can be assigned for each line extension)
- Busy Lamp Field (BLF)
- · Distinctive incoming call treatment / call waiting
- · Call timer
- · Call transfer, hold, divert (forward), pickup
- · Called, calling, connected party information
- · Local three-way conferencing
- One-touch speed dial, redial
- · Call waiting
- · Remote missed call notification
- Intercom
- · Automatic off-hook call placement
- Do not disturb function

#### Other Features

- · Local feature-rich GUI
- Time and date display
- User-configurable contact directory and call history (missed, placed, and received)
- Customisable call progress tones
- Wave file support for call progress tones
- Unicode UTF-8 character support. Multilingual user interface encompassing Chinese, Danish, Dutch, English (Canada / US / UK), French, German, Italian, Japanese, Korean, Norwegian, Portuguese, Russian, Spanish, Swedish

#### **Protocol Support**

• IETF SIP (RFC 3261 and companion RFCs)

### Network and Provisioning

- Two-port Gigabit Ethernet switch 10/100/1000Base-Tx across LAN and PC ports Conforms to IEEE802.3-2005 (Clause 40) for Physical Media Attachment - Conforms to IEEE802.3-2002 (Clause 28) for Link Partner Auto-Negotiation
- Manual or dynamic host configuration protocol (DHCP) network setup
- Time and date synchronisation using SNTP
- FTP / TFTP / HTTP / HTTPS serverbased central provisioning for mass deployments
- Provisioning and call server redundancy supported
- Web portal for individual unit configuration
- QoS Support IEEE 802.1p/Q tagging (VLAN), Layer 3 TOS, and DSCP
- Network Address Translation (NAT) support for static configuration and "Keep-Alive" SIP signalling
- RTCP support (RFC 1889)
- Event logging
- Syslog
- Local digit map

- · Hardware diagnostics
- · Status and statistics reporting

### Security

- Transport Layer Security (TLS)
- Secure Real-time Transport Protocol (SRTP)
- Shipped with X.509 certificate installed
- · 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
- External Source 48V DC @ 380mA

#### Approvals

- FCC Part 15 (CFR 47) Class B
- ICES-003 Class B
- EN55022 Class B
- CISPR22 Class B
- AS/NZS CISPR 22 Class B
- VCCI Class B
- EN55024
- EN61000-3-2; EN61000-3-3
- · ROHS compliant
- Anatel

#### Safety

- CE Mark
- EN 60950-1
- IEC 60950-1
- AS/NZS 60950-1
- UL/C-UL

### **Operating Conditions**

- Temperature: 0 to 40°C (+32 to 104°F)
- Relative Humidity: 5% to 95%, non-condensing

#### Storage Temperature

• -40 to +70°C (-40 to +160°F)

### SoundPoint IP 560 comes with:

- SoundPoint IP 560 console
- · Handset with handset cord
- Base stand

- Network (LAN) cable
- Quick Start Guide
- · Product registration card

#### Size

• 26.5 cm x 15 cm x 19 cm x 6.5 cm) (W x H x D x T) (10.5 in x 6 in x 7.5 in x 2.5 in)

### Weight

- Phone weight: 1.1 kg (2.4 lb) Unit Box Dimensions / Weight
- 31.75cm x 33.66cm x 8.89cm (12.5 in x 13.25 in x 3.5 in)
- 1.8 kg. (4.0 lbs)

### Master Carton Quantity

• Five

### Country of Origin

Thailand

### Warranty

• 1 Year

Price: £214.30

Options available for Polycom SoundPoint IP 560 VoIP Phone (IP560):

### **Power Supply Required?**

- Not Required -, Require Polycom Power Supply (+£10.50).