

ErisStation™ SIP Conference Phone with Four Wireless Mics

Includes 2 fixed microphones and 4 DECT wireless microphones with Orbitlink Wireless Technology™ and support for up to 3 SIP lines/accounts

VCS754A

Whether you're already running your business with a SIP phone system or getting ready to make the move, you can extend your coverage all the way to the conference room with the VTech ErisStation SIP conference phone. It comes with four removable DECT microphones that feature Orbitlink Wireless Technology™. Unlike traditional conference phones, ErisStation lets everyone spread out and speak without having to gather around the phone or raise their voices. Advanced audio features, such as acoustic echo cancellation, automatic gain control and dynamic noise reduction, deliver premium, interference-free sound quality. Plus, integrated design makes it easy to charge the microphones right in the base and keep everything together.



ErisStation SIP Conference Phone Features¹

Central, full-duplex speakerphone in base

Callers can speak and be heard at the same time, just as they would in a face-to-face meeting.

Flexibility

With support for up to three SIP lines/accounts, you can manage multiple calls with advanced call handling features.

Intuitive design

You're in business with a user-friendly interface, large backlit display, three soft keys and navigational pad.

USB speakerphone

USB connection for using a PC-based softphone or speaker.

HD voice quality

With G.722 wideband codec support for HD audio, calls have never sounded so rich and life-like.

DECT wireless microphones with Orbitlink Wireless Technology™

Portable, wireless microphones allow everyone to collaborate beyond the conference table.

Up to 8 hours of talk time per wireless microphone

With up to 8 hours of talk time for each microphone, you have plenty of room to keep the meeting going on one charge.

Up to 70 hours of standby time per wireless microphone

With up to 70 hours of standby time for each microphone, there's no need to worry about losing a charge mid-meeting.

Built-in wireless microphone charging bays in base

Magnetic charging bays in the base keep the microphones together and ready for the next call.

Operable in small, medium and large rooms

With a base speakerphone, 2 fixed and four wireless microphones, ErisStation is a flexible solution for any size conference room.

Automatic gain control

- Digital mixing
- Volume control: 10 levels

3-way local conference

N-way network conference

Local phonebook up to 200 entries

Call history up to 200 entries

ErisStation SIP Conference Phone Specifications, Weights & Dimensions

Color: GM (Gun Metal)	Giftbox weight: 2.51kgs
Product dimensions (WxDxH): BS: 273mm x 342mm x 67.1mm	Master pack dimensions (WxDxH): 238mm x 328mm x 420mm
Product weight: 1.76kgs	Master pack weight: 5.67kgs
Giftbox dimensions (WxDxH): 314mm x 395mm x 110mm	Quantity: 2

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ErisStation SIP Conference Phone Specifications

Physical Features

- 158 x 57 pixels (w x h) LCD resolution with backlight (white)
- 3 soft keys with 4-way navigational pad
- Built-in wireless microphone charging bays in base
- Includes 2 fixed microphones and 4 DECT wireless microphones with Orbitlink Wireless Technology
- USB connection for PC softphone/speakerphone
- Active call indicator; green LED on base unit and wireless microphones
- Mute indicator; red LED on base unit and wireless microphones
- Wireless microphone battery-low indicator; flashing LED
- Wireless microphone auto power off; when base powered off
- Operable in small, medium and large rooms
- AC adapter required (included)

Phone Features

- Supports up to 3 SIP account registrations
- 10 ring tones/melodies; 10 ring levels
- 10 speed dial entries
- Pre-dialing with editing
- Local phone book up to 200 entries
- Call history up to 200 entries
- Alpha phone book search
- Do Not Disturb (DND)
- 3-way local conference
- N-way network conference
- Call barring/blocking (block anonymous/blacklist)
- Dial plan/digit map
- Call transfer

- Call hold
- Call forwarding: per-line basis (busy/always/no answer)
- Call timer (call duration)
- Phonebook matching (incoming and outgoing calls)

Audio / Voice

- Central, full-duplex speakerphone in base
- Fast adaptive feedback control
- Acoustic echo cancellation technology
- Comfort noise injection
- Enhanced noise reduction/Dynamic noise reduction (DNR)
- Automatic gain control
- Digital mixing
- Volume control: 10 levels

Battery / Charging

- Up to 70 hours of standby time per wireless microphone
- Up to 8 hours of talk time per wireless microphone

Security

- 802.1x
- 802.1pq
- Transport Layer Security (TLS)
- AES encryption of config file for provisioning (128/64 / 32 / 16 bits)
- LLDP-MED

Network

- STUN
- NAT

- RTP, SRTP
- SNTP
- VLAN
- Layer 3 QoS (ToS, Diffserv)

Voice Codec

- G.711 a-law & u-law
- G.722
- G.726
- G.729

Provisioning

- HTTP, HTTPS, FTP, TFTP
- DHCP options: 66, 159,160
- Web interface configuration (admin & user)
- PnP

Maintenance

- Diagnostic: pcap trace
- Syslog

Compatibility

- Broadsoft® Broadworks
- Asterisk-based platforms
- Metaswitch Networks
- Genband™

Other

- Eco mode (talk mode)

¹ Features and functionality are subject to change without notice and may vary, depending on SIP service provider and/or SIP phone system platform.