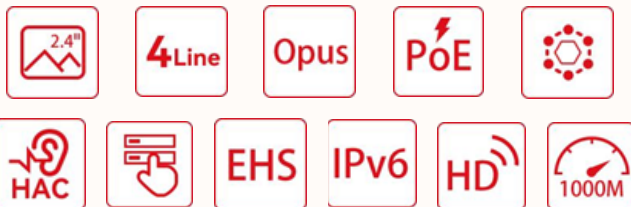




WIP-10 (G/W)



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Prime Business IP Phone

Wevetel WIP10G is a stylish and user-friendly IP phone. It features a 2.4-inch (320x240) high-definition color screen, supports 4 SIP lines, and allows for a 6-way audio conference. With rich functions such as HD audio, dual Gigabit ports, and integrated PoE, the WIP10G is suitable for various business application scenarios, providing users with an excellent interactive experience and comfortable instant communication services.

Next-Generation Communication Solutions

- ◆ 2.4-inch 320x240 pixel color screen.
- ◆ Support audio codec G.722 and Opus, HD audio on speakerphone and handset.
- ◆ Support 6-way audio conference.
- ◆ Available to link with security products, can be easily customized into various security solutions for reducing investment.
- ◆ Dual Gigabit ports, integrated PoE.
- ◆ Up to 12 DSS keys, every DSS key can be set as Line/BLF/Speed Dial Key.

Generic

- 2.4-inch (320x240) color screen
- 4 SIP lines
- HD voice
- Dual Gigabit ports, PoE enabled
- Intelligent DSS keys
- Handset / hands-free / headset mode
- Desktop Stand / Wall-mounted (optional)
- Optional external power supply
- Handset with Hall-effect Sensor Switch

Phone Functions

- Local Phonebook (1000 entries)
- Remote Phonebook (XML/LDAP)
- Call logs (In/out/missed, 1000 entries)
- Blocked List/Allowed List
- Screen saver
- Voice Message Waiting Indication (VMWI)
- Programmable DSS/Soft keys
- Network Time Synchronization
- Support Poly wireless headset (Through Poly APD-80 EHS Cable)
- Support Jabra wireless headset (Through Wevetel EHS20 EHS Cable)
- Support Recording (Through Server)
- Action URL / Active URI
- XML Browser
- Customize Background
- Customize Boot Logo
- uaCSTA
- Customize Incoming Call Ringtone
- Time Plan (timed reboot/timed forward/timed upgrade/timed config)
- Hidden DTMF
- Ring From Headset
- Customize DSS icon
- Customize DSS LED lights
- Customize BLF
- Customize UI Display

Call Functions

- Call out / Answer / Reject
- Mute / Unmute (Microphone)
- Call Hold / Resume/Call Waiting
- Intercom
- Caller ID Display
- Speed Dial
- Anonymous Call (Hide CallerID)/ Ban Anonymous call
- Call Forward (Always/Busy/No Answer)
- Call Transfer (Attended/Unattended)
- Call Parking/Pick-up (Depending on server)
- Redial/Predial
- Call Back
- Do-Not-Disturb(DND)/Timed DND
- Auto-Answering
- SIP SMS
- Voice Message(On server)
- 6-way Audio Conference
- Hot Line
- Hot Desking
- Direct IP call without SIP proxy
- Fuzzy Search Call
- Action Plan
- Dial Plan
- Intercom
- Call Completion/Auto Redial
- Primary and Backup SIP Servers
- Default Ext Line
- Multicast

Audio

- HD Voice:HD handset, HD speaker
- HAC handset
- Wideband ADC/DAC16KHz Sampling
- Narrowband Codec: G.711a/u, G.723.1, G.726, G.729,G.729A, G.729B, G.729AB,iLBC
- Wideband Codec:G.722, Opus
- Full-duplex AcousticEcho Cancellor (AEC)
- Voice Activity Detection (VAD) / Comfort Noise Generation (CNG) / Background Noise Estimation (BNE) / Noise Reduction (NR) / Automatic Gain Control (AGC)
- Packet Loss Concealment (PLC)

- Dynamic AdaptiveJitter Buffer
- DTMF: In-band, Out-of-Band (RFC2833/ SIP INFO)
- VQM voice quality monitoring

Network

- Physical: 10/100/1000 Mbps Ethernet, dual bridged port for PC bypass
- IP Mode: IPv4/IPv6/IPv4&IPv6
- IP Configuration: Static IP / DHCP / PPPoE
- Network AccessControl: 802.1x
- VPN: L2TP / OpenVPN
- VLAN
- LLDP
- QoS
- RTCP-XR (RFC3611), VQ-RTCPXR (RFC6035)
- DNS-SRV (RFC 3263)
- Support Redundant Servers
- Agent
- Https Mutual Authentication

Protocols

- SIP v1 (RFC2543) v2(RFC3261)
- UDP/TCP/TLS
- RTP/RTCP/SRTP (Optional/Compulsory)
- STUN
- DHCP
- IPv6
- LLDP
- PPPoE
- 802.1x
- 802.1p/Q
- L2TP
- OpenVPN
- SNTP
- FTP/TFTP
- HTTP/HTTPS
- AES128 & AES256
- TR-069
- AES128 & AES256

Deployment & Management

- Auto-Provisioning via FTP/TFTP/ HTTP/HTTPS/DHCP OPT66/SIP PNP/TR-069
- Screenshot
- I Web Management Portal

- Web-based Packet Dump
- Configuration Export/ Import
- Phonebook Import/Export
- Firmware Upgrade
- Syslog
- Network Firewall

Physical Specifications

- LCD: 2.4-inch(320x240) Color Screen
- Keypad: 35 keys, including
 - 3 DSS keys + 1 page key (with LED)
 - 5 Function keys (Contact, MWI, Mute, Headset, Redial)
 - 4 Soft-keys
 - 4 Navigation keys, 1 OK key
 - 1 Return key, 1 Home key
 - 12 Standard Phone Digits keys
 - 2 Volume Control keys, Up/Down
 - 1 Hands-free key
- Handset (RJ9)x1
- Standard RJ9 Handset Wire x1
- 1.5M CAT.5E Ethernet Cable x1
- Desktop Standx1
- Status Indicator Light x1 (Red)
- RJ9 Port x2: Handsetx1, Headset x1
- RJ45 Port x2: Networkx1, PC x1 (Bridged to Network)
- Security Slot x1
- Power: DC5V/1Aor PoE
 - POE complieswith the IEEE 802.3af standard
 - Power adapter:
 - Input: 100-240VAC~50/60Hz
 - Output:5V/1A
- Power Consumption (Adapter): 1.33~3.00W
- Power Consumption (PoE): 1.90~4.60W
- Working Temperature: 0~45°C
- Working Humidity: 10~95%
- Installation:
 - Desktop Stand
 - Wall mountable (optional)
- Color: Dark Grey
- Qty/CTN: 10 PCS
- N.W/CTN: 9.56 kg
- G.W/CTN: 10.26 kg
- Device Dimensions:
 - Desktop Stand(40°): 191.8 x 164.5 x 174.2 mm
 - Desktop Stand(45°): 191.8 x 154.5 x 181.9 mm
 - Wall mountable: 191.8 x 91.6 x 215.6 mm
- Gift Box Dimensions: 295 x 222 x 56 mm
- Outer CTN Dimensions: 464 x 305 x 315 mm (10 PCS)