

Entry-level Business IP Phone UT-S1E/P

Highlights:

- Simple and fashionable appearance of the product
- Economic and Environmental Friendly design with Elegant housing.
- Easy installation and configuration
- Compatible with major platforms: 3CX, Asterisk, Broadsoft, Elastix, etc.



The New UT-S1E(P) IP Phone is a basic phone with the advanced design. Packed with all the necessary VoIP features at an affordable price, it is the best choice for customers who only need essential VoIP access.

Generic

- 2 SIP Line
- POE Enabled(S1EP Only)
- Handset(HS) / Hands-free(HF) / Headphone(HP) mode
- Desktop stand
- Optional External Power Supply

Call Features

- Call out / answer / reject
- Mute / Unmute (microphone)
- Call Hold / Resume
- Call Waiting
- Intercom
- Caller ID Display
- Speed Dial
- Anonymous Call (Hide Caller ID)
- Call Forwarding (Always/Busy/No Answer)
- Call Transfer(Attended/Unattended)
- Call Parking / Pick-up (depending on server)
- Redial/Auto-Redial
- Do-Not-Disturb (per line / per phone)
- Auto-Answering (per line)
- Voice Message (on server)
- Local 3-way Conference
- Hot Line
- Hot-Desking

Phone Features

- Phonebook (500 entries)
- Remote Phonebook (XML/LDAP)
- Call log (600 entries, in/out/missed)
- Black/White List Call Filtering
- Voice Message Waiting Indication (VMWI)
- Programmable Soft keys
- Network Time Synchronization
- Action URL / Active URI
- uaCSTA

Audio

- Narrowband CODEC: G.711a/u, G.722, G.726-32K, G.729A, Opus, iLBC
- Full-duplex Acoustic Echo Canceller (AEC) – Hands-free Mode, 96ms tail-length
- Voice Activity Detection (VAD) / Comfort Noise Generation (CNG) / Background Noise Estimation (BNE) / Noise Reduction (NR)
- Packet Loss Concealment (PLC)
- Dynamic Adaptive Jitter Buffer up to 300ms
- DTMF: In-band, Out-of-Band – DTMF-Relay(RFC2833) / SIP INFO

Network Features

- 10/100Mbps Ethernet, dual bridged port for PC bypass
- IP Configuration: Static / DHCP / PPPoE
- IPv4, IPv6, IPv4&IPv6
- Network Access Control: 802.1x
- VPN: L2TP (Basic Unencrypted) / OpenVPN
- VLAN
- QoS

Maintenance & Management

- Auto-Provisioning via FTP/TFTP/HTTP/HTTPS/DHCP OPT66/SIP PNP/TR069
- Web Management Portal
- Web-based Packet-dump
- Configuration Export / Import
- Phonebook Import/Export
- Firmware Upgrade
- Syslog

Physical Features

- LCD: 128x48 dot-matrix, no backlight
- Keypad: 24 keys, including
- 4 Soft-keys
- 4 Function keys (Up, Down, Headset, Redial)
- 12 Standard Phone Digits keys
- 2 Volume Control keys, Up/Down
- 1 Hands-free key
- 1 Mute key
- Standard RJ9 Handset Wire x1
- 1.5M CAT5 Ethernet Cable x1
- RJ9 Phone port x2: Handset x1, Headphone x1
- RJ45 Port x2: Network x1, PC x1 (Bridged to Network)
- DC Power Input: 5V / 0.6A
- Power Consumption: Idle 0.89W~1.2W Peak 1.81W~2.5W
- Working Temperature: 0~40°C
- Working Humidity: 10~95%
- Color: Black
- Device Dimensions: 153x165.8x163.1mm
- Gift Box Dimensions: 250x205x66mm
- Outer CTN Dimensions: 420x340x265mm (10 PCS)