

HD Entry Level Phone UC902













IPPHONETHE FUTURE



Phone Features

2 SIP accounts
Call hold, mute, DND
One-touch speed dial, hotline
Call forward, call waiting, call transfer
Redial, call return, auto answer, direct IP call
5-way conferencing
Group listening, SMS, emergency call
Ring tone selection/provisioning
Set date time automatically or manually
Dial plan for each account
RTCP-XR (RFC3611), VQ-RTCPXR (RFC6035)
Action URL/UR

Audio Features

HD Voice: HD Speaker, HD Handset , HD codec Codecs: iLBC, G.722, G.711(A/µ), Opus,G.723.1,

G.729AB,G.726-32

DTMF: In-band, RFC 2833, SIP INFO

Full-duplex hands-free speakerphone with AEC

AEC, VAD, AGC, CNG, PLC, AJB

Directory

Local phonebook up to 1000 entries XML/LDAP remote phonebook Intelligent search method Phonebook search/import/export Call history: dialed/received/missed/forwarded Black list

IP-PBX Features

Busy Lamp Field (BLF), bridged Line Appearance(BL A)

Anonymous call, anonymous call rejection
Message Waiting Indicator (MWI), voice mail
Intercom, paging
Hot-desking
Call park, call pickup
Music on hold

Display and Indicator

132x48 -pixel graphical LCD with backlight LED for call and message waiting indication National language selection Caller ID with name, number

Feature keys

2 line keys with LED

6 feature keys: mute, headset, message transfer, redial, speaker

4 context -sensitive "soft" keys

6 Navigation keys(include cancel key) and volume control keys

Interface

Dual -port 10/100M Ethernet Power over Ethernet (IEEE 802.3af), class 2 1xRJ9 (4P4C) handset port 1xRJ9 (4P4C) headset port

Physical Features

Wall mountable
Stand with 2 adjustable angles
External universal AC adapter:
AC 100~240V input and DC 5V/1.2A output
Power consumption (PSU): 1.6~2.6W
Power consumption (PoE): 2.0~3.2W
Operating humidity: 10~95%
Operating temperature: -10~50°C

Management

Configuration: browser/LCD -Menu/auto -provision Auto provision via HTTP/HTTPS FTP/TFTP Auto -provision with PnP Reset to factory, restart, reboot Local tracing log export, system log Phone lock for personal privacy protection

Network and Security

SIP v1 (RFC2543), v2 (RFC3261)
SIP server/proxy redundancy
NAT Traversal: STUN mode
IP assignment: static/DHCP
HTTP/HTTPS web server
Time and date synchronization using SNTP
DNS-NAPTR/DNS-SRV(RFC 3263)
QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
Transport Layer Security(TLS1.2), SRTP, ZRTP
HTTPS certificate manager
AES encryption for configuration file
Digest authentication using MD5/MD5-sess
IEEE802.1X/OpenVPN
IPv4/IPv6, LLDP/CDP/DHCP VLAN





