Ubiq Com

Ubiqcom P-52





Model: P-52

Display	Display Resolution: 128x48	
Phone Features	 Supports SIP2.0 (RFC3261) and Corelative RFCs Supports 2 SIP servers Supports SIP UDP/ TCP/ TLS SIP authentication (none, basic, MD5) Peer to Peer / IP Call 	
Feature Keys	Large dot matrix LCD displayProgrammable Soft keys & Function keys	
Audio Features	 HD voice: HD codec Codec: G.711A/ u, G.723 high/low, G.729a/b, G.722, G.726 DTMF: In-band, Out - of – band (RFC 2833) and SIP INFO Full duplex hands- free speaker phone Voice Activity Detection (VAD) Comfort Noise Generation (CNG), Ech cancellation: Support G.168, Hand free Speaker Phone 	10
Network Features	 WAN / LAN : 10/100M Ethernet ports Support PPPoE for xDSL, POE (optional) Support VLAN, 802.1X Support basic NAT and NAPT NAT Transverse: Support STUN client Support DNS Relay, SNTP Client Firewall, VPN (L2TP/Open VPN) and DMZ, Network tool in telnet Server: Including ping, trace route, Telnet Client 	
Management Features	 WEB, Telnet and keypad management Upgrade firmware through POST mode and HTTP, FTP or TFTP Support DHCP option66 and Custom Option, Auto-Provisioning Telnet remote management/ upload/download setting file Safe mode provide reliability Support Auto Provisioning to upgrade Firmware or configuration file with HTT Support TR-069 and Syslog 	ΓPS
Physical Features	 Adapter Input :100-240V Adapter Output : 5V/1A WAN/LAN Port -10/100 BASE-T Power: 2.5W/Active: 2.8W operation Temperature:0 ~40 °C Relative Humidity 10 ~65 % Main Chipset - Broadcom SDRAM - 16MB, FLASH 4MB Weight - Phone: 0.84kg Dimension: 250x205x60 mm 	
Advanced Features	 Call forwarding, Call Hold, Call Waiting Call transfer (blind/ attended/ alert) Call Conference, Auto Redial Call Pickup, Call Completion Support messaging and MWI Flexible dial plan Barring function for outgoing calls Do not disturb, Auto answer, Caller ID display Call forwarding, Call Hold, Call Dial without registration Call Logs: 300 records Support SMS and Speed Dial Hotline/Warm-line, Black List Intercom/Intercom barge Support action URL/URI Voice codec setting for each SIP line Hands-free ringing choice Support XML phonebook/browser Support Click to Dial 	

Certification







