

ES220 is a highly innovative based VoIP phone, It can perfectly satisfy with all kinds of the businessman communication. ES220 integrates with 132x48 graphic LCD, a large resolution display screen, elegant and intuitive user interface, extend adjustable bracket, which indicate you can enjoy perfect user experience. ES220 offers high audio quality HD handset, HD speaker and HD Code (G.722), with a rich set of business communication features that can definitely improve working efficiency and sense. ES220 is highly suitable for SMEs, Home Office and ISP applications.

Key Features

- Extend adjustable bracket.
- HD Voice: HD Codec, HD Speaker, HD Handset.
- Convenient and fast for Installation and maintenance Support HTTP/TFTP/Telnet/Auto provision, and TR069 centralization maintenance.
- Multi-language, e.g. Chinese, English, Russian, France etc.
- Configuration for speed dial, hotline etc.
- Two SIP accounts and support three-way conference, SMS.
- Enterprise Phonebook, LDAP Phonebook



ES220 Technical Datasheet

Phone Features

2 SIP accounts, Hotline Call hold, Call waiting, Call forward, Call return Call transfer (blind/busy/ask) Caller ID display, Redial, Mute, DND Auto-answer, 3-way conferencing Speed dial, SMS, Voicemail Message Waiting Indication LED Tone scheme, Volume control Direct IP call without SIP proxy Ring tone selection/import/delete Black list, Hand-free indicator. Call history: dialed/received/missed (50 -entries); Multi-Language Soft keys programmable, Supports PC control

Advanced Features

XML Phone Book search and input & output Enterprise phone book(800 entries) Personal Phonebook(300 entries) LDAP phone book Functions customizable keys

IP PBX System Integration

Busy lamp field (BLF), BLF list Bridged line appearance (BLA) DND & Forward synchronization Intercom, Paging, Music on hold Call pickup, Dial Plan, Call recording Anonymous call, Anonymous call rejection

Compatible with the platform

Codes and Voice Features

Wideband Codec: G.722 Narrowband codec: G.711µ/A, G.723.1 G.726, G.729AB,iLBC VAD, CNG, AEC, AGC, Full-duplex, Security 802.1x, VLAN QoS (802.1pq), SRTP* Transport Layer Security (TLS)* Digest authentication using MD5/MD5-sess Secure configuration file via AES encryption* Encryption for configuration file* Phone lock for personal privacy protection Admin/User 2-level configuration mode , LLDP, CDP L2TP VPN tunneling protocol - added

Network Features

SIP v1 (RFC2543), v2 (RFC3261) SIP connection: Proxy mode & Peering Points mode; IPV6 DNS SRV (RFC3263); SRTP voice encryption(RFC3711) Redundant server support NAT Traversal: STUN mode DTMF: In-Band, RFC2833, SIP Info, Auto HTTP/HTTPS Web Management IP Assignment: Static/DHCP/PPPoE Bridge/Router mode for PC port TFTP/DHCP/PPPoE client Telnet/HTTP/HTTPS server, DNS client, NAT/DHCP server Layer 3 ToS, DSCP Management

Auto-provision via FTP/TFTP/HTTP/HTTPS SNMP V1/2, TR069 optional Configuration: browser/phone/auto-provision Trace package and system log export



132x48 graphic LCD 27 keys, including 2 account keys 6 LEDs: 1*status light, 1*speaker light 2* accounts lights,1* message light,1* headset light. 1* RJ-9 handset port 2xRJ45 10/100M Ethernet ports Extend bracket Power adapter: AC 100~240V input and DC 5V/1A output Power over Ethernet ,IEEE 802.3af,class 0 Power consumption: 2.5-3.5W Net weight: 1.07KG Dimension: 206x166x59MM Operating humidity: 10~95% Storage temperature: up to 60°C

Package Features

Qty/CTN: 10 PCS N.W/CTN: 10.7KG G.W/CTN: 13.94KG Carton Meas: 485*460*300MM

Certifications





ESCENE COMMUNICATION CO., LTD

Add: Floor 4 building C YueAn Industrial Park No.51 Huangcun Road Tianhe District, Guangzhou, China. Tel: (86)20-82320720

*these features are not available and blocked in firmware for products which are supplied to Russian Federation and EAC countries