

Escene ES292N is a highly innovative based SIP VoIP phone, It can perfectly satisfy with all kinds of the businessman communication. ES292N integrates with 132x64 graphic LCD, a large resolution display screen, elegant and intuitive user interface, which indicate you can enjoy perfect user experience.ES292N offers high audio quality HD handset, HD speaker and HD codec (G.722), with a rich set of business communication features that can definitely improve working efficiency and sense. Escene ES292N is highly suitable for SMEs, Home Office and ISP applications.

Key Features

- 132*64 Graphic LCD.
- HD Voice: HD Codec, HD Speaker, HD Handset.
- Multi-language: Chinese, English, French, Russian, etc.
- 2 level adjustable bracket and wall-mountable.
- Three SIP accounts and support three-way conference, SMS.
- 2xLAN, PoE(option), 3.5mm and RJ9Headset.
- USB port only support DC 5V output feature.
- XML/LDA, Contact Search.
- Auto-provision, HTTP/TFTP/FTP, Tr069.

ES292N Technical Datasheet

Phone Features

3 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

Phonebook (Personal Phone Book

Search/Input/Output (300 entries),

Black list, Hand-free indicator.

Call history: dialed/received/missed (100

-entries); One-key back (C)

Multi-Language (2aB/abc/ABC/123)

Soft keys programmable, Supports PC control

Support switching SIP server & Second server

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

Network conference, Distinctive ringtone

Security

802.1x, VLAN QoS (802.1pq)*

Transport Layer Security (TLS)*

Digest authentication using MD5/MD5-sess

Secure configuration file via AES encryption*

Phone lock for personal privacy protection

Admin /User 2-level configuration mode

Network Features

SIP v1 (RFC2543), v2 (RFC3261)

SIP connection: Proxy mode & Peering Points

DNS SRV (RFC3263)

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

Management

Auto-provision via FTP/TFTP/HTTP/HTTPS

SNMP V1/2, TR069

Configuration: browser/phone/auto-provision

Trace package and system log export

Codes and Voice Features

Wideband Codec: G.722

Narrowband codec: G.711µ/A, G.723.1

G.726, G.729AB, VAD, CNG, AEC, AGC

Full-duplex,

Physical Features

132x64 graphic LCD

34 keys including 3 programmable keys

1x3.5mm handset port,1xRJ9 headset port

2xRJ45 10/100M Ethernet ports

2 level adjustable bracket and wall-mountable

Power adapter: AC 100~240V input and

DC 5V/1A output

USB port only support DC 5V output feature.

Power over Ethernet ,IEEE 802.3af,class 0

Power consumption: 1.4-2.6W

Net weight: 0.80KG

Dimension: 196×39×204MM

Operating humidity: 10~95%

Storage temperature: up to 60°C

Package Features

Qty/CTN: 10 PCS

N.W/CTN:12.50 KGS

G.W/CTN: 13.50KGS

Measurement: 0.068CBM(0.5% more or less)

Carton Meas: 285×220×90MM

Certifications









Compatible with the platform





























Communication starts from SayHi



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