

VoiSmart MINI IP PBX



Product Name: VoiSmart MINI IP PBX

Manufacturer: -

Model Number: MINI IP

Availability: In Stock

VoiSmart Mini IP PBX

Mini IP manages up to 16 users and 6 nominal lines with a combination VoIP and integrated ISDN BRI. Mini IP is built with anodized aluminum compact fanless case, it uses only flash memory in order to avoid any movement parts. Mini IP has integrated 2 ISDN BRI interfaces, 2 Ethernet interfaces, 2 USB and a serial port for debugging. Target: SOHO, small office, remote office Mini IP is small in size but big in performances, as it offers the same high end software that VoiSmart deploys in its large corporate Unified Communications systems: the power of a giant in a nutshell!

- Ideal up to 16 users
- Maximum 6 lines (VoIP and/or PSTN)
- Analog, ISDN interfaces with external gateways
- Terminals IP, BCA (with gateway), WiFi
- Case fanless/diskless, compact flash, 2 USB ports, 2 LAN 10/100 ports
- Fax Server Integrated up to 2 concurrent faxes (optional)

VoiSmart is a manufacturer of telephone systems, not a licensing software maker. Together with a mini PBX, an full set of telephone services is supplied without any additional cost:

- Intuitive and user friendly web interface
- Voicemail
- Unified communication suite
- Fully customisable auto attendant
- Customisable music on hold
- PC operator panel
- Automatic Call Distribution
- Integrated phone book
- Programmable Least Cost Routing on real cost
- Audio conference agenda
- Authentication service on active directories and LDAP
- Click and call
- Calls user and queues statistics and graphs

These services can be activated, through a graphical web user interface, by the system administrator; no extra cost for any additional user/ip phone/terminals. Call functions

- Call waiting
- Caller ID identification
- Call waiting indication (caller ID on call waiting)
- Caller name display
- Caller ID Blocking
- Call forwarding on busy
- Call forwarding on no answer
- Call forwarding to other extension, group, operator
- Call forwarding based on caller ID
- Call transfer blind
- Call transfer assisted
- Call transfer parking
- Call transfer retrieval
- Call waiting management

- Remote Call pickup
- Call-hunting services
- Call-back on busy
- Do Not Disturb, DND function
- Call-hunting multilevel
- Call back function
- Director-secretary services
- Speed dialling
- Music on Hold, Music on Transfer
- Music and messages: Flexible MP3 based system, Volume Control, Random Play, Linear Play

Protocol

- SIP
- Traditional Telephony Interoperability, FXS,FXO, BRI, PRI GSM Protocols
- DTMF support

Codec ADPCM, G.711 (A-Law & μ -Law), G.722, G.723.1 (pass through), G.726, G.729 (optional), GSM, iLBC, Linear, LPC-10, Speex Management

- Web based interface, intuitive point-and-click, drop-down menus, online help
- multiplatform solution
- multilingual interface
- Remote assistant using VPN
- Graphic Dial Plan editor
- Autoprovisioning IP phones
- Alarms with e-mail notify and Automatic Backup
- Internal DHCP Server
- Internal Firewall

Telephony Services

- VoiceMail for accounts, password protection, call notification, lost calls and voicemail presence indication
- Access to VoiceMail through phone, browser, e-mail, web and SMS(optional)
- Customisable voice messages
- Fully customisable auto attendant (day, night, weekend, holidays)
- Fully programmable IVR platform
- Flexible extensions numbering
- Multilines extensions
- DISA (Direct Inward System Access)
- Multiple audio conference rooms
- Unlimited audio conference users (limited by the specific IP PBX nominal lines)
- LDAP directory services
- Local phonebooks: Public, Personal, Groups, Extensions
- Shared phonebooks: LDAP, remote PBXs
- Fully programmable call queueing functions (round robin, ring all, random, least recent, fewest calls)
- Call Details Record (access based on privileges)
- Call Details Record by lines and telecom operator
- System statistics and graphs
- Calls user and queues statistics and graphs
- Calls logging (missed calls, dialed calls)
- ACD (Automatic Call Distribution)
- Remote and local Agents support

- Authentication service on active directories and LDAP
- Intercom service on ip phones
- Programmable multiple Least Cost Routing on real cost
- Recording call suite (access based on privileges)
- Distinctive ringing (depending on ip phone)
- Remote phonebooks on ip phones (depending on ip phone model)
- Operatore panel and Presence on PC (web or stand-alone)
- speed dial phonebook
- Text-to-Speech
- Computer-Telephony Integration

Please Enquire

VoIPVN <http://voip.com.vn>