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VST3300 Series SIP IP-PBX



Product Overview

The Edge-Core VST3300 Series are highly versatile, integrating Analog Trunk, Analog Line, SIP Trunk, SIP Line and IP-PBX with in one box. With Built-in Registrar and Proxy Server, the VST3300 series can serve point to multi-point branches without ISP services and an extra telephone fees. The VST3300 series can also work with existing PBXs to extend current telephone networks. This all-in-one VoIP device is the best solution for multi-point, small group connections.

Key Features and Benefits

Integrate Analog Trunk, Analog Line, SIP Trunk, SIP Server, IP-PBX into 1 device There is no need to purchase several devices for a complete solution. Complete trunking is available using analog PSTN ports and the SIP trunk. Comprehensive extension numbers are available by employing Analog and SIP line ports.

Built-in Registrar and Proxy Server support SIP standard devices RFC-3261 and RFC-2833 SIP-compliant devices, such as SIP Gateways and IP Phones, can register as an extension to the VST330.

Scalable up to 100 sites and 8800 extensions Great adaptability and system planning are provided with 2, 4, 8, 16 analog ports and scalability, which when set, works as a complete IP-PBX.

Register to ITSPs (IP telephony Service Providers) for additional, value-added service. Reduce communication costs with the additional ability to register to IP telecoms

Original dialing behavior. Dialing plans can be maintained with E.164, offering high flexibility.

Multiple "Call Forwarding" Never miss a call. The VST3300 can ring a variety of devices, such as GSM mobile phones, notebooks, IP Phones, etc in an order pre-arranged by the user. Call Forwarding can be set for busy, no answer, and all calls

Private IP support giving SIP devices "Plug and Play" mobility. Public IP is not required. The VST3300 uses Media Relay to reach devices behind NAT without additional configuration.

Upgrade software through Web interface.Users have simple, on-demand maintenance for configuration files, auto attendant greetings and software using this convenient feature.

Functions

Built-in DISA provides recorded auto attendant messages and transfers incoming calls. Messages can be easily recorded through the telephone or downloaded by FTP.

Network Operator can be assigned to any extension and be reached by dialing "0" when calling from the PSTN.

Inbound, Outbound and Transit Calls are all supported. Even calls from the PSTN can be transited to other SIP numbers via the SIP trunk

Call transfer, call park and consult transfer to any extension of local or remote sites through Analog or SIP lines.

Call forward an incoming call to an analog line, a SIP line of a local or remote VST3300 or to a PSTN number.

Automatic Trunk Selection chooses the cheapest routing for outbound calls.

Call Pick-up isolates incoming calls into an area where other extensions belonging to the same pick-up group can retrieve it.

Speed dialing can be configured for 100 abbreviate numbers.

T.38 FAX is supported to connect a fax machine to an analog port of the VST3300.

Life Line allows calls to be made though the analog line port when the power is off.

Comprehensive Networking Management through telephone set, Console, SNMP and WEB management and Telnet. Moreover, the VST3300 is manageable from remote sites.

CDR Collection, or Call Detail Record, through the RS-232 port or TCP/IP protocol for better call accounting management.



VST3300 Product Specifications

Features

System Scale

Maximum Networking: 100

Maximum Extensions: Analog: 1600 / SIP: 7200 Maximum Trunk: Analog: 1600 / SIP: 1600

Data/Voice Management

Protocols

- RTP (Transfer Protocol)
- Proprietary SIP (Call Control Protocol)
- Standard SIP Protocol (RFC-3261, RFC-2833)

Voice Codec: G.711 / 20ms, G.723 / 60ms, G.729 / 40ms Voice Quality

- VAD (Voice Activity Detection)
- CNG (Comfortable Noise Generation)
- Echo Cancellation (G.165/G.168 16ms)
- Adaptive Jitter Buffer Management

Gain Control: In/Out +/-6db

- IP Address Support
- Static IP
- DHCP
- Private IP

Mechanical

Power Input:

100 ~ 240VAC 50 ~ 60Hz

EXS

SIP Line

External Power Adapter

Emissions/Safety

Safety: cUL, CCC*, CB* *Future Release PTT Regulations: FCC part 68, JATE*, CTR21 EMI Certifications: FCC Part 15 Class B, CE Mark

Environmental Specifications

Working Environment

- -10°C to 70°C (Storage Temperature)
- 0°C to 50°C (Operating Temperature)

SIP PBX Features

Auto Attendant (DISA)

Music on Hold

Network Operator Group

Barring Class Trunk Group

Trunk Class

Line Group

Automatic Trunk Selection Specific Trunk Selection

Network Extension Line

Inbound Transit Call

Outbound Transit call Least Call Routing

Traffic Management

Call Transfer

Call Forward

Call Pick Up group

Call Park

Abbreviated Dialing DND (Do Not Disturb)

Life Line

T.38 FAX

Management

Management:

- WEB Browser
- Phone Set ■ System Console
- Telnet

Software Upgrade: ■ FTP

- WEB
- Multiple configuration files
 - Configuration file upload/download via TFTP server

Power

Consumption

Concurrent

Calls

Limited Lifetime Warranty

Ordering	Informat	ion
Models	EYO	

VST3305	1	2	8 / 72*	2 / 24*	36	10W	190mm x 25mm x 110mm
VST3306A	2	2	12 / 72*	2 / 24*	36	12W	172mm x 35mm x 176mm
VST3306B	0	4	12 / 72*	2 / 24*	36	12W	172mm x 35mm x 176mm
VST3306C	4	0	12 / 72*	2 / 24*	36	12W	172mm x 35mm x 176mm
VST3318A	8	8	32/72*	4 / 24*	48	70W	440mm x 66mm x 254mm
VST3318D	4	12	32 / 72*	4 / 24*	48	70W	440mm x 66mm x 254mm
VST3318E	12	4	32 / 72*	4 / 24*	48	70W	440mm x 66mm x 254mm

SIP

*Maximum Amount Supported for Optional SIP License Expansions

Optional Accessories

Modules:

MP3008

MP3108

MP3208

SIP License Upgrade Keys:

SIPL-4

SIPL-8

SIPT-2

SIPT-4

d g e - c o r

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8 FXS ports 8 FXO ports

Product Description

4 FXS + 4 FXO ports

Upgrade Key for 4 SIP Lines

Upgrade Key for 8 SIP Lines

Upgrade Key for 2 SIP Trunks

Upgrade Key for 4 SIP Trunks

Dimensions