

VoxStack 3G Gateway

There are three models with VoxStack series of 3G Gateway, the VS-GW1202, VS-GW1600 and VS-GW2120. The Modular Design of 3G Gateways are ranging from 4 up to 44 3G/UMTS channels, it is designed for multiple frequency bands Including 3G/UMTS 850/900/1900/2100 MHz, and it can support a wide selection of codecs, including G.711A, G.711U, G.729, G.722, G.726. For a reliable connecting 3G to voip gateway, it can quickly reduce telecommunication expenses and maximize cost-savings. With the unique design of the VoxStack Gateway, it can support hot-swap for both SIM cards and 3G gateway modules. Users can simply add or remove the modules for hardware expansion or exchange. Each 3G gateway module runs an independent asterisk system inside. The VoxStack gateway designs with two LAN Switch boards that provide stackability on the hardware upgrade. The 3G Gateways will be 100% compatible with asterisk, Elastix, trixbox, 3CX, FreeSWITCH SIP server and VOS VoIP operating platform.



Appearance







VS-GW1202-8W



VS-GW2120-28W

Parameter

•	Size: VS-GW1600	44cm*4.5cm*30cm	•	Weight: VS-GW1600	3682g
	VS-GW1202	15cm*19cm*4.5cm		VS-GW1202	502g
	VS-GW2120	44cm*34cm*9cm		VS-GW2120	6144g



LAN Port: 2

SIM Cards: Hot-Swap

GWM400G Module: Hot-Swap

Operation Humidity Range:10%~90%

non-condensing

Frequency Bands: UC15-E

900/2100MHz@UMTS 900/1800MHz@GSM

UC15-A

850/1900MHz@UMTS

850/900/1800/1900MHz@GSM

UC15-T

850/2100MHz@UMTS

850/900/1800/1900MHz@GSM

Storage Temperature Range: -40°C ~85°C

Operation Temperature Range: -40°C ~85°C

Power: VS-GW1600-20W 46W VS-GW1202-8W

> VS-GW2120 95W

18W

3G Features

- Volume Adjustment
- CLID Display & Hide (Need operators' support)
- Real Open API Protocol (based on Asterisk)
- Echo Cancellation (G.168)
- Call Duration Limitation
- SMSC/SMS/USSD
- SMS Bulk Transceiver, Sent to Email and Automatically Resent

- Gain Adjustment
- PIN Identification
- **IMEI Number Automatically Modify**
- **Band Binding**
- **Call Transfer**
- **Polarity Reversal**
- Adaptive (dynamic) Jitter Buffer
- Call Forwarding

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7*24 Technical Support

Any Way You Like



SIP Characters

- Add, Modify & Delete SIP Trunk
- SIP Registration with Domain
- SIP V2.0 RFC3261
- SDP RFC2327
- Session Timer RFC4028
- RTP/RTCP RFC3551
- DNS SRV/A Query
 - Network
- IPv4,IPv6(Coming soon),UDP/TCP, PPPo
 E, DHCP, TFTP
- IPv6 (Coming soon)

- Combine Different SIP Trunks into Group
- Multiple SIP Registrations: Anonymous, Endpoint registers with this gateway (work as a SIP Sever), This gateway registers with the endpoint.
- DTMF Mode: RFC2833/Inband/Info
- SIP, IAX, TCP, UDP, RTP, TELNET, HTTP, SMTP, POP3
- HTTP/SSH (Optical Telnet)
- Ping & Traceroute Command on the Web
- Two Types of IP Access

V

System Features

- Abundant Codecs : G.711A, G.711U,
 G.729, G.722, G.723, G.726, GSM
- Simple and convenient configuration via Web GUI
- Firmware Update by HTTP
- Automatically Reboot
- Extensible Automatic Callback and Speed
 Dial
- Without Physical Serial (Provide Virtual Serial via TCP/IP Protocol)

- Language Selection (Chinese/English)
- Multiple Detailed LOG Output
- Call Status Display
- PDD/ACD/ASR/BER Display
- CDR (More than 200,000 Lines CDRs Storage Locally)
- Restore Factory Settings
- Customizable IVR/DISA
- High Equipment Materials Specifications,
 Suitable for Long Distance Transportation

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