



Open Vox DGW-L20X Series E1T1PRI VoIP Gateway

Open Vox DGW-L20X T1/E1 Gateway with open-source Asterisk-based VoIP Gateway solution is an operator and call center. It is a converged media gateway product. This type of gateway connects traditional phone systems to IP networks and smoothly incorporates VoIP PBX and ISDN. With a friendly GUI, users may easily set up their customized Gateway. Also, secondary development can be through AMI (Asterisk Management Interface). The DGW-L20X T1/E1 Gateway helps 1/2/4 software-selectable T1/E1/PRI interface manage up to 120 continuous calls.



Technical Specifications

- | | |
|--|---|
| <ul style="list-style-type: none">◆ 1/2/4 T1/E1 RJ-48◆ 2 10/100/1000M Ethernet ports◆ 2 USB 2.0 ports◆ DGW-100XR with redundant power supply◆ Maximum Power Consumption: 20W | <ul style="list-style-type: none">◆ Power supply specification: 100-240V/AC◆ Operation humidity: 5%~95% non-condensing◆ Operating temperature: 0°C~70°C◆ Storage temperature: -40°C~85°C |
|--|---|

<i>Features</i>	
<ul style="list-style-type: none"> ◆ Available in 1/2/4 port T1/E1, energy efficiency concurrent processing, upto 120 ◆ Signal ling: PRI/R2/SS7 ◆ Support up to 24 countries' standard R2 signal ling ◆ Support new R2 variant ◆ Simple and convenient configuration via Web GUI ◆ Codecs support: G.711A, G.711U, G.729A, G.723.1, G.722, GSM ◆ SIP(RFC3261) compliance ◆ Independent System for Each Module ◆ Customizable IVR 	<ul style="list-style-type: none"> ◆ Affordable Price with Superior Performance ◆ Hot-swap Design for both Sim cards and Modules ◆ Compatible with sip server, such as: Asterisk, Elastix, 3CX, Free SWITCH, etc. ◆ Available for OEM ◆ 3-Month "No Questions Asked" Return Policy ◆ One Year Warranty ◆ PIN identification ◆ Mobile number portability (MNP) ◆ Flexible routing settings & unlimited routing rules

<i>SIP Features</i>	
<ul style="list-style-type: none"> ◆ Support add, modify & delete SIP Accounts ◆ SIP registration with Domain ◆ Support multiple SIP registrations: Anonymous, Endpoint registers with this gateway and this gateway registers with the endpoint 	<ul style="list-style-type: none"> ◆ SIP accounts can be registered to multiple servers ◆ Combine different SIP Trunks into group ◆ SIP(RFC3261) compliance ◆ DTMF: RFC2833, SIP INFO, INBAND ◆ Support T.38 /Pass-through Fax

<i>Routing</i>	
<ul style="list-style-type: none"> ◆ Flexible routing settings ◆ Support 512 routing ◆ Support caller/callee manipulation and filtering 	<ul style="list-style-type: none"> ◆ Trunk group support, Trunk priority management ◆ Support add, modify & delete routing ◆ E1/T1 port grouping ◆ Support Failover

Network Features	
<ul style="list-style-type: none"> ◆ Network type: Static IP and DHCP ◆ IPv4, UDP/TCP, DHCP, TFTP, SCP ◆ HTTP/HTTPS/SSH 	<ul style="list-style-type: none"> ◆ Support DDNS ◆ Support ping & trace route command on the web ◆ Support network capture on the web

OpenVox DGW-L20X Series E1T1PRI VoIP Gateway			
Module Type	GSM	WCDMA	LTE
Band	850/900/1800/1900M Hz@GSM	900/2100MHz @UMTS 900/1800MHz @GSM	LTE FDD: B1/B3/B5/B8 LTE TDD: B38/B38/B40/B41 WCDMA: B1/B8 TD-SCDMA: B34/B39 CDMA: BC0 GSM: 900/1800MHz

Office Address	Contact Us Via Mobile	Contact Us Via Mail
----- Asfera Technologies Private Limited , Saroj Tower, Office No:- 15 First Floor, 59/1, above Bank of Baroda, Kalkaji Extension, New Delhi, 110019.	----- +91-9066677770 +91-1146579777	----- info@asfera.in sales@asfera.in