



### OpenVox DGW-L1 Series E1T1PRI VoIP Gateway

OpenVox DGW-L1 T1/E1 Gateway is an open-source Asterisk-based VoIP Gateway solution for operators and call centers. It is a converged media gateway product. With a friendly GUI, users may easily set up their customized. Also, secondary development can be through AMI (Asterisk Management Interface). The DGW-L1 T1/E1 Gateway helps one software-selectable T1/E1/PRI interface with supports up to 30 concurrent calls.



### *Technical Specifications*

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| <ul style="list-style-type: none"><li>◆ 1/2/4 T1/E1 RJ-48</li><li>◆ 2 10/100/1000M Ethernet ports</li><li>◆ 2 USB 2.0 ports</li><li>◆ DGW-100XR with redundant power supply</li><li>◆ Maximum Power Consumption: 20W</li></ul> | <ul style="list-style-type: none"><li>◆ Power supply specification: 100-240V/AC</li><li>◆ Operation humidity: 5%~95% non-condensing</li><li>◆ Operating temperature: 0°C~70°C</li><li>◆ Storage temperature: -40°C~85°C</li></ul> |
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<i>Features</i>	
<ul style="list-style-type: none"> <li>◆ Available in 1/2/4 port T1/E1, energy efficiency concurrent processing, upto 120</li> <li>◆ Signal ling: PRI/R2/SS7</li> <li>◆ Support up to 24 countries' standard R2 signal ling</li> <li>◆ Support new R2 variant</li> <li>◆ Simple and convenient configuration via Web GUI</li> <li>◆ Codecs support: G.711A, G.711U, G.729A, G.723.1, G.722, GSM</li> <li>◆ SIP(RFC3261) compliance</li> <li>◆ Independent System for Each Module</li> <li>◆ Customizable IVR</li> </ul>	<ul style="list-style-type: none"> <li>◆ Affordable Price with Superior Performance</li> <li>◆ Hot-swap Design for both Sim cards and Modules</li> <li>◆ Compatible with sip server, such as: Asterisk, Elastix, 3CX, Free SWITCH, etc.</li> <li>◆ Available for OEM</li> <li>◆ 3-Month "No Questions Asked" Return Policy</li> <li>◆ One Year Warranty</li> <li>◆ PIN identification</li> <li>◆ Mobile number portability (MNP)</li> <li>◆ Flexible routing settings &amp; unlimited routing rules</li> </ul>

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

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

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

<b>OpenVox DGW-L1 Series E1T1PRI VoIP Gateway</b>			
Module Type	GSM	WCDMA	LTE
Band	850/900/1800/1900MHz@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

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