Created on: Friday 08 August, 2025

PORTech MV-372L - 2 channel GSM/VoIP Gateway - 4G LTE

Product Name: PORTech MV-372L - 2 channel GSM/VoIP Gateway - 4G LTE
Manufacturer: PORTech Communications
Model Number: MV-372L

Portech MV-372L - Great Value 2-Channel Gateway - 4G LTE

Please Note: This product may have a lead time of up to 1-2 weeks. Please call 0330 088 0195 and speak to our sales team who can confirm the current available stock for your requirement

The PORTech MV-372 is a dual channel VoIP GSM Gateway which allows for call termination (VoIP to GSM) and call origination (GSM to VoIP.) The gateway fully supports Asterisk, it connects as a SIP trunk and allows you take advantage of low cost calls to mobile phones via the SIM card.
Key Features:

ï¿½ VoIP (SIP) - GSM conversion
ï¿½ GSM - VoIP conversion
ï¿½ 2 Simultaneous GSM calls (with 2 SIMs inserted)
ï¿½ Space for 50 mobile toLAN route settings
ï¿½ Space for 50 LAN to mobile route settings
ï¿½ Voice response for setting and status (dial in from mobile)
ï¿½ Standard SIP protocol
ï¿½ Full web browser configuration
ï¿½ Send and receive SMS
ï¿½ Quad-band

The MV-372 supports 2 SIM cards so supports two GSM channels at a time. It allows the users to make two simultaneous calls from IP phones to GSM or GSM to IP phones. The PORTech MV-372 is a quad-band device, which should work in most territories worldwide.
The MV-372 can receive calls from the user (via mobile or landline) and then forward the call via the Internet to an IP PBX, VoIP gateway or ITSP (such as ourselves) and then onto SIP phones, analog phones, PSTN or a mobile phone.
Simple 2-stage dialling:- The user dials the number of the SIM thats inside the MV-372, the gateway then presents the user with a dial tone, and DTMF signalling is then used to pick up the desired destination. The MV-372 then routes the call via GSM or VoIP depending on the settings that have been configured. All settings can be configured via a Web interface.
When an IP phone and the MV-372 both register to the SIP proxy Server or Asterisk server, you can dial any destination number from IP phone directly.

Specification

ï¿½ Protocols: SIP (RFC2543, RFC3261)
ï¿½ TCP/IP: IP/TCP/UDP/RTP/RTCP, CMP/ARP/RARP/SNTP, DHCP/DNS Client , IEEE802.1P/Q, ToS/Diffserv, NAT Traversal, STUN, uPnP, IP Assignment, Static IP, DHCP, PPPoE
ï¿½ Codec: G.711 u-Law/a-Law, G.723.1 (5.3k), G.723.1 (6.3k), G.729A, G.729A/B, Voice Quality, VAD, CNG, AEC, LEC, Packet loss
ï¿½ GSM: Quad-band: 850/900/1800/1900 MHz

Price: $516.87