



Product Name: Positron Telecom V-600 Asterisk PCI Card

Manufacturer: -Model Number: V600

Availability: In Stock

Positron V-600 Quad E1 T1 PCI Express Card

High density enterprise communication solution for voice, fax and videoThe V-600 from Positron Telecom is a powerful and flexible enterprise solution capable of supporting a full function PBX, T.30 fax termination on all ports, T.38 fax termination, voice and video trans-coding – all on a single board solution. The V-600 installs simply and quickly into your system as an Ethernet adapter and will accept SIP traffic without the need for application development or additional software – just configure and go. Technical Specifications

� PCle (1x)

ï¿⅓ Four E1/T1 PRI ports

� Dedicated hardware echo cancellers per E1/T1

� Powerful on board CPU @ 1.3Ghz

ī స్ట్రీ On board multicore DSP for media processing (15 cores)

ï¿.1/2 SATA 2.0 controller

าั¿½ 10/100/1000 Mbps Ethernet for PCI and RJ-45

Highlights

าั¿½ Installs as Ethernet adapter – just send SIP traffic, no development required

ï¿⅓ No CPU load on Host system

ï¿1/2 No Host driver required

ï¿⅓ Installs in ANY operating system

آذِ1/2 Ideal for Hosted VMWARE environments

ï¿⅓ Integrated PBX)

� Unlimited boards per Host system

� Cross Board Switching

Applications Voice Transcoding Gateway 900 G.729 to G.711 simultaneous

� Adaptive Multi Rate (AMR)

� AMR-WB (G.722.2)

� Enhanced Full Rate (GSM-EFR)

آز¼ G 711

ï¿⅓ G.723.1

� G.726

ï¿⅓ G.729AB

ï¿1/2 G.729.1

73½ G.722.1 نظر

� iLBC

ï¿⅓ Speex

Video Transcoding Gateway

ï¿1/2 Video Support- H.263

� CIF/QCIF - MPEG-4 Part-2

T¿½ CIF/QCIF- H.264 encoder (demo purposes) - Transcoding - Resizing - Frame rate adaptation- Bit-rate adaptation

آز½ Video Effects

ï¿⅓ Video mixer



- � Text overlay
- ï¿⅓ Graphic Overlay

IP PBX with integrated Echo Cancelling and telephony ports

- ï¿⅓ Tone Detection / Generation
- ï¿⅓ Tone Relay
- ï¿⅓ Caller ID Detection / Generation
- ï¿1/2 Conferencing
- � V.152
- ï¿1/2 SIP to TDM Gateway
- SRTP ½'sï

Fax Server

- � 200 T.38 Sessions over UDP or RTP
- � 120 T.30 simultaneous faxs (V.17 at 14.4Kbps)

Audio Conferencing Server

- ï¿1/2 Adaptive Noise Reduction
- ï¿1/2 Automatic Level Control
- ï¿1/2 High Level Compensation
- � Natural Level Enhancement

PBX FunctionalityExtensions

- آذِ1/2 IP Phone / ATA extensions
- � Virtual Extensions
- ï¿⅓ IVRs
- � Extension Templates
- � Control permissions for each extension
- ïز½ 3,4,5 digit extensions

Mobility Features

- ïز½ Find me / Follow me
- � Voice / Fax / Email to inbox
- � Corporate callback
- ï¿1/2 Caller ID on call forward

Calling Methods

- ۲۶⅓ VoIP گئï
- � T1/E1 Phone Lines
- ï¿1/2 Connecting Multiple PCI cards

Call Control

- ۲۶½ Hold
- � Assisted Transfer
- � Blind Transfer
- تز½ Call Parking
- � Do Not Disturb
- ï¿1/2 Send Calls



- � Call forwarding rules
- � Direct Inward dial (DID)
- ï¿⅓ Call Waiting
- � Called ID
- � Call Parking
- � Call retrieve / pickup

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Conferencing

- ï¿1/2 Simple Conference Room
- ï¿1/2 Meet Me Conference Center
- ï¿1/2 Three way conference

Music On Hold

- ï¿1/2 Custom Music on Hold
- ï¿1/2 Music on Hold included

Recording & Donitoring Call Recording Logging & Donitoring Call Recording Call Recording Logging & Donitoring Call Recording Call Re

- � Current Calls
- � Call Logs
- ï¿1/2 Call Reporting

Auto attendant

- � Dial By Extension Directory
- � Dial By Name Directory

More Features

- ï¿⅓ Advanced Diagnostics
- � Least cost routing
- ï¿⅓ Failover between PSTN and VoIP
- ï¿1/2 Auto phone provisioning
- ï¿⅓ Status page monitor
- � Desktop paging (needs phone support)
- � Multilingual support



Price: £2,298.00