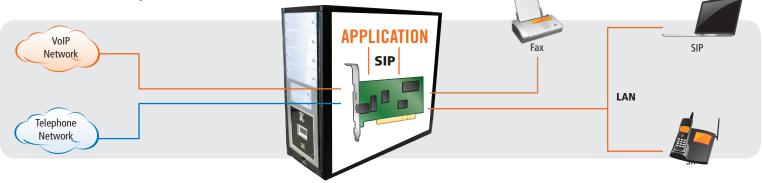
POSITRONtelecom

V-310 PCI

The age of unified communication is upon us. Positron Telecom offers the V-Series family of cards, which will change the way business and households communicate. The V-Series provides PBX and telephony ports which creates a seamless gateway to the cellular, traditional telephone and VoIP worlds by combining them into a single integrated device.

An affordable, scalable solution for single point of contact communication needs, the V-310 PCI enables customers to communicate through VoIP or telephone lines, provides a centralized communication point and routes calls accordingly to desk, home or cellular phone. The V-Series also provides customized greetings per user, lower cost long distance and a true one inbox solution for email, fax and voicemail – essential for today's communication needs.



PCI CARD | GATEWAY | ECHO CANCELLATION | PBX | SIP/IAX

THE V-310 PCI CARD IS A **FULLY FUNCTIONAL PBX** ON A SINGLE PCI CARD WITH INTEGRATED:

- Hardware echo cancellation
- Ethernet WAN port

- Single-port connection for a phone or fax (FXS)
- Single E1/T1/PRI port for digital telephone lines

Unlike traditional telephony adapters, the V-Series is a standalone system that provides both telephony and PBX on the card itself. The V-Series offers a unique approach because it installs as an Ethernet adapter into the system automatically providing an IP address for easy configuration. This allows the card to be easily installed into any operating system, providing a simple way for card-to-card communication for higher density installs.

SCALABLE

- 4 VoIP lines
- Single E1/T1/PRI interface

POWERFUL

- Hardware echo cancellation
- High performance CPU

SIMPLE

- Web-based configuration
- Single board solution

SAVE

- Long distance using VoIP
- No more extra charges for voicemail

Any Operating System

The V-Series is installed as an Ethernet adapter and is automatically assigned an IP address

Simple Configuration

Quick and easy setup and configuration is achieved through a web-based configuration interface

Auto Configure SIP Phone

An integrated module seamlessly detects and installs SIP phones

Remote User Capability

Allows users to connect through PBX with SIP phones even when working remotely

Free Voicemail

The system accepts, stores and converts all voicemails and sends them to email

Reduce Long Distance Charges

Benefit from very low VoIP long distance rate

Integrated VoIP and Analog

Seamless integration of VoIP and analog telephones into the same system

THINK INSIDE THE BOX



Key Features & Benefits

Analog and Digital Phone Lines

 Ability to transparently support standard phone lines and VoIP lines. Offers the advantage of having additional low cost telephone lines without tying up the main line

Low Cost Long Distance VolP Calls

- Unlimited VoIP phone numbers
- Customizable greetings and features for multiple telephone lines

PBX Features

- PBX features include: call hold, call transfer, call forward, caller ID, call waiting
- Advanced features such as converting voicemail to email, conferencing and Findme

Low Cost Calls From Around the World

- Purchase an International DID
 (phone number) from your VoIP service
 provider (very low monthly cost).
 Allows "termination" of the phone number
 into the PBX. Our system allows
 configuration of unlimited inbound
 accounts. For example:
 - Purchase a NY DID (area code 212) to deal with a contact in NY or an Ireland DID (area code 01) to deal with a contact in Ireland. This will allow them to call you from NY or Ireland using a local number

Voicemail to Email

- All users can have voicemail directly sent to email as a WAV or MP3 file (audio attachment). Voicemails can then be retrieved from the inbox on a computer or mobile phone
- Messages can be heard through an integrated media player rather than logging into the system
- Allows for a true one inbox solution for emails and voicemails

Customized Settings per User

- All settings are configurable for each user. For example:
 - Customized extension numbers: 2201 or 2
 - Customizable voicemail for each user

VoIP Failover

 Calls can still be made through the regular phone line in the event of Internet connection failure

Long Distance Calls from a Mobile Phone

 Ability to make long distance calls from a mobile phone by calling the system, entering a password and hanging up.
The system will call back and prompt for a number to call. Enter the number on the mobile phone and the system will call that number and join the calls together. This provides the advantage of making a long distance call at the rate of an inbound local call

Find Me

- Ability to set up call forwarding rules on a per-user basis that will help people stay in touch. For example:
 - From Monday to Friday set the calls to forward to a mobile phone between 8:00 am and 6:00 pm
 - If the central number is not answered after four rings, forward to a mobile phone

Call Pick-up

- Ability to answer any calls, whether on hold or ringing, even if the call is coming from another line For example:
 - If the phone rings in a different department or in a different room in a house, just dial *99 and the call will be picked up

Parental Control

 Ability to log, record and transparently listen to phone calls

Digital Receptionist

- Dial-by-name directory feature
- Easily create custom greetings:
 - "Press 1 for sales, 2 for accounting, 3 for marketing, ..."
 - "Hi and welcome to the Smiths. Dial 1 for mom, 2 for Dad, 3 for Sarah and 4 for Lucy"





E1/T1 1 Port

FXS 1 Port **VolP**Savings

SIP Protoco

10/100 Ethernet **QOS** Quality

