

digium

TE122
TE120P/TE122



User

601-00010

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Note: Canada, Finland, Norway, Sweden and the United States of America require that equipment using this product must be a Restricted Access Location (RAL).

Telecom:

FCC Part 68, ANSI/ITA-968-A, Including Amendment A

Industry Canada CS-03

AS/ACIF S016: 2001

AS/ACIF S038: 2001

TBR4 November 1995 as amended by TBR4/A1 December 1995

TBR12 December 1993

TBR13 January 1996

EMC:

EN 55022:1998 Class B and 47 CFR Part 15, Subpart B Class B Limits for Radiated and Conducted EN 55024:1998 / IEC 61000

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A plug and jack used to connect this equipment to the pre and telephone network must comply with the applicable rules and requirements adopted by the ACTA.

If your TE120 Series card causes harm to the telephone network, the telephone company may notify you in advance that temporary discontinuance of service may be required. But if advance notice is not practical, the telephone company will notify you as soon as possible. Also, you will be advised of your right to file a complaint with the FCC if you believe it is necessary.

The telephone company may make changes in its facilities, equipment, operations or procedures that could affect the operation of this equipment. If this happens, the telephone company will provide advance notice in order for you to make necessary modifications to maintain service.

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If you experience problems with the TE120 Series, contact Technical Support +1.256.428.6161 for repair and/or warranty information. If the equipment is causing harm to the telephone company, the telephone company may request that you disconnect the equipment until the problem is resolved.

FCC Part 15

This device complies with part 15 of FCC rules. Operation is subject to the following two conditions: (1) This device may not cause interference, and (2) This device must accept any interference, including interference that may cause undesired operation.

Industry Canada Compliance Information

The Industry Canada label applied to the product (identifying the Industry Canada logo or the "IC:" in front of the certification number) indicates that the Industry Canada technical specifications are met.

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| Chapter/ Appendix | Title | Description |
|----------------------|-----------------------|---|
| 1 | Overview | Identifies your card’s features. The card covers applications and uses for the card in the real world. |
| 2 | Card Installation | Provides instructions for installing the card on a PC, acquiring correct drivers, and ensuring compatibility. |
| 3 | Configuration | Provides instructions for configuring the card. |
| 4 | Troubleshooting | Explains resolutions to common problems and frequently asked questions pertaining to installation and usage. |
| A | Pin Assignments | Lists the connectors and pin assignments for the card. |
| B | Specifications | Details card specifications. |
| C | Glossary and Acronyms | A list of terms and acronyms used in this manual. |

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electrostatic discharge (ESD) wrist strap while han



The Electrical Hazard Symbol indicates a possibility of electrical shock when operating this unit in certain situations. To avoid risk of damage or injury, follow all steps or procedures as instructed.

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Water and Moisture.
Do not spill liquids on this unit. Do not operate this unit in a wet environment.



Heat.
Do not operate or store this product near heat sources, radiators, air ducts, areas subject to direct, intense sunlight, or other products that produce heat.



Static Electricity.
To reduce the risk of damaging the unit or your equipment, do not attempt to open the enclosure or gain access to areas not instructed to do so. Refer servicing to qualified service personnel.

Save these instructions for future reference

Service Personnel Cautions



Warning.
This card must be used with the PC lid screwed down. Telecommunications network voltages exist inside the PC. The PC must be shut down and telecommunications equipment shall be removed before opening the PC.



Electrical Shock.
To reduce the risk of injury, damage to the unit or your equipment, do not attempt to touch the modules while they are powered. The unit should be securely closed before power is applied to the unit.

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equipment previously labelled **Permitted** or **Certified**



Caution.

Only connect regulatory equipment (approved for use in your country) to the telecommunications network voltage



Caution.

This card is not intended for home use. It must be used at authorized access locations and installed in UL Listed I.T.E. only

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What is Asterisk®?
Asterisk as a Switch (PBX)
Asterisk as a Gateway
Asterisk in the Call Center
Asterisk in the Network
Asterisk Everywhere

Chapter 2

Card Installation
 Unpacking the Card
 Shipment Inspection
 Identifying Features
 T1/E1 Selection
 Slot Compatibility
 Hardware Installation
 Software Installation
 Installing Asterisk

Chapter 3

Configuration
 Configuring Card Features

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Pin Assignments

Appendix B

Specifications

Appendix C

Glossary and Acronyms

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Robbed Bit Signaling (also known as CAS or Channel Associated Signaling), CCS (Common Channel Signaling), E&M, PRI, ISDN (PRI), and several data modes (PPP, HDLC, Cisco frame relay). The TE120 Series cards are capable of running J1 modes.

Designed to be fully compatible with existing software applications, integrate fully with the Asterisk platform, the TE120 Series offers many advanced call features.

Data Modes:

- Cisco HDLC
- HDLC
- PPP
- Multilink PPP
- Frame Relay

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- 5ESS (Lucent)
- DMS100
- Q.SIG

- E&M

- Wink
- Feature Group B
- Feature Group D

- FXO and FXS

- Ground Start
- Loop Start
- Loop Start with Disconnect Detect

The TE120 Series cards can be used to connect your Asterisk to the PSTN world, your channel bank, or even another PBX, all accomplished via a T1/E1 interface. The cards allow Asterisk to connect to your network, creating a professional telephony environment. Figure 1 and Figure 2 show examples of the card's applications.

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Figure 1: Sample Legacy Phone Application

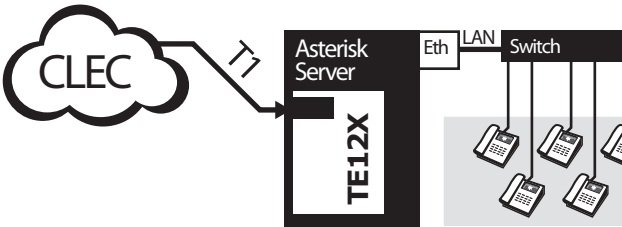


Figure 2: Sample IP Phone Application

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The TE120 Series cards, unless otherwise equipped, utilize Asterisk mainline software to perform software-based echo cancellation. Asterisk mainline software is a collection of open source echo cancelers. These open source echo cancelers provide a moderate level of echo cancellation, but are not capable of handling higher levels of, or more advanced, echoes.

Digium recommends that those users concerned about echo cancellation should purchase the VPMADT032 hardware echo cancellation module. The VPMADT032 may be combined with both the TE121 and TE122B cards, but may not be combined with the TE120P card. The TE121 and TE122B are offered bundled with the VPMADT032 as, respectively: TE121B and TE122B.

The VPMADT032 is designed to handle up to 128ms of echo in both directions across all channels and provides a G.168 compliant echo cancellation solution.

If enabled and not explicitly disabled in chan_dahdi.conf, the VPMADT032 will automatically operate and cancel all echoes within its tail range (1024 taps). Users of TE120P cards, however, may maintain the capability to support the VPMADT032, may also purchase Digium's commercial HPEC software:

<http://www.digium.com/en/products/software/hpec.php>

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for download free of charge. Asterisk is the most popular software available, with the Asterisk Community being the largest in VoIP.

Asterisk as a Switch (PBX)

Asterisk can be configured as the core of an IP or hybrid PBX, handling calls, managing routes, enabling features, and connecting to the outside world over IP, analog (POTS), and digital (T1/E1) lines.

Asterisk runs on a wide variety of operating systems including Linux, Mac OS X, OpenBSD, FreeBSD, and Sun Solaris. It provides all the features you would expect from a PBX including many advanced features that are often associated with high end (and high cost) proprietary PBXs. Asterisk's architecture is designed for maximum flexibility, supporting Voice over IP in many protocols, and can interoperate with existing standards-based telephony equipment using relatively inexpensive hardware.

Asterisk as a Gateway

It can also be built out as the heart of a media gateway, bridging the legacy PSTN to the expanding world of IP telephony. Asterisk's architecture allows it to convert between a wide range of protocols and media codecs.

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Asterisk in the Call Center

Asterisk has been adopted by call centers around the world for its flexibility. Call center and contact center developers have implemented ACD systems based on Asterisk. Asterisk has also added to existing call center solutions by adding remote IP agent capabilities, advanced skills-based routing, predictive and bulk dialing.

Asterisk in the Network

Internet Telephony Service Providers (ITSPs), competitive local exchange carriers (CLECS) and even first-tier incumbent local exchange carriers discovered the power of open source communications with Asterisk. Feature servers, hosted services clusters, voicemail systems and other calling solutions, all based on Asterisk have helped reduce costs and enabled flexibility.

Asterisk Everywhere

Asterisk has become the basis for thousands of communications solutions. If you need to communicate, Asterisk is your answer. For more information on Asterisk visit <http://www.asterisk.org> or <http://www.digium.com>.

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- **Shipment Inspection** on page 22
- **Identifying Features** on page 22
- **T1/E1 Selection** on page 22
- **Slot Compatibility** on page 26
- **Hardware Installation** on page 28
- **Software Installation** on page 29
- **Installing Asterisk** on page 35

Note: The TE120 Series card installation instructions that they will apply to any card in the series. Example specific information are included as needed.

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shipment.

Note: Only qualified service personnel should install this equipment. Users should not attempt to perform this function themselves. The equipment must ensure that the equipment is permanently connected to a power source of a pluggable type B or connected to a socket-outlet that has a protective earth to ensure that it is reliably earthed in accordance with the applicable National Electrical Code.



This card is intended for installation in a Restricted Access Location (RAL) only.

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Your TE120 Series card has one RJ45 port and two status LEDs: a green status LED and an amber loop-back LED. The card includes a jumper for selecting either T1 or E1 line mode. See Figure 3 on page 24 for details of these features.

The TE121 and TE122 cards may also be combined with a hardware-based echo canceler, model VPMADT032. See page 24 for an example of the TE121 card shown with the cancellation module.

T1/E1 Selection

The T1/E1 mode, in most cases, is set at the distributor before shipment. You may want to check the setting to be certain it is set for your use. With the jumper **off**, the span is ready for T1 mode, and with the jumper **on**, the span is ready for E1 mode.

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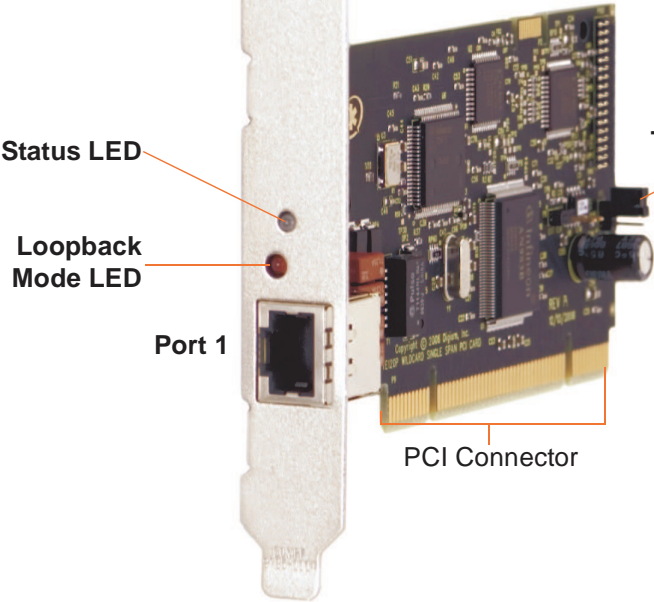


Figure 3: TE120P Card

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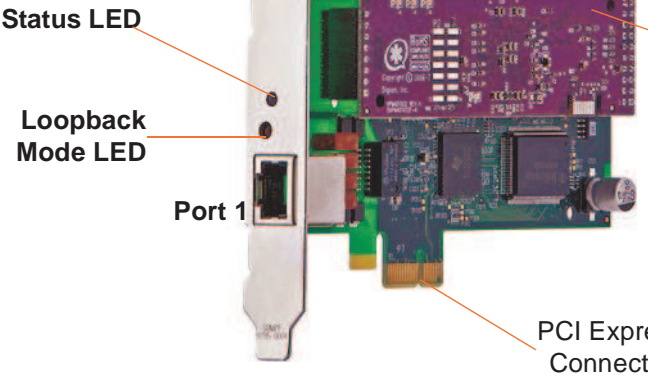


Figure 4: TE121 Card with Echo Cancellation

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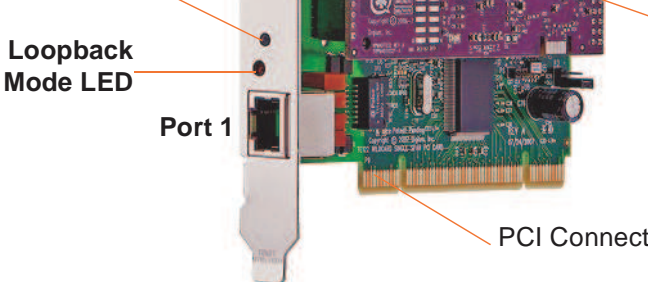


Figure 5: TE122 Card with Echo Cancellation



Caution.
Only qualified service personnel should continue hardware installation and configuration of the TE122 card. Non-qualified personnel should not attempt these functions themselves.

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- 1: 64-bit 5.0 volt PCI Slot
- 2: 64-bit 3.3 volt PCI Slot
- 3: 32-bit 5.0 volt PCI Slot
- 4: PCI Express X1 Slot

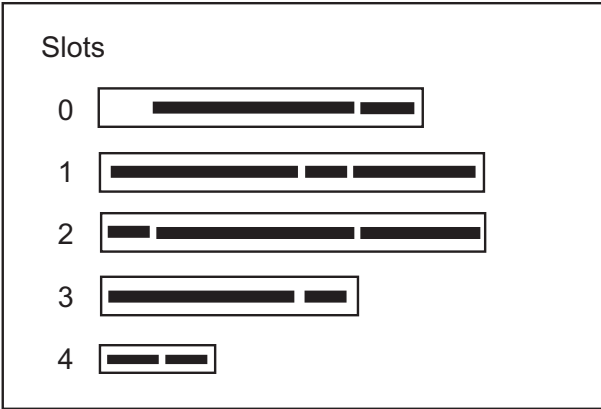


Figure 6: Motherboard Slots

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(X1) PCI Express compliant slot. The TE121 will work in any PCI Express compliant slot, including lane lengths X1, X4, X8 and X16. This means that in the motherboard shown in Figure 6, the TE121 can be inserted into Slot 4. The TE121 **can not** be used in Slots 0 through 3.

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ports for E1. Setting the jumper with the jumper strap
ports for T1.

4. Remove the bracket place holder and insert the card in Express slot. See Figure 7 for an example of card installation.

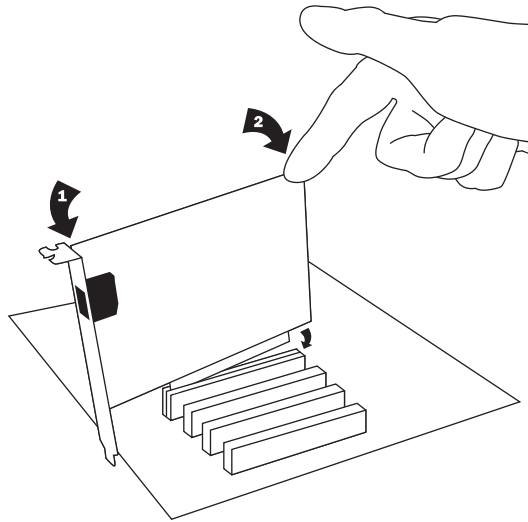


Figure 7: Insert the Card

5. Replace the cover to your computer.

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**Caution.**

This unit must be connected only to the appropriate Telecommunications Network port (as approved for the specific country).

Software Installation

Digium hardware requires drivers and libraries that are not included with the Linux kernel. Digium hardware is only supported on Linux. Digium recommends CentOS, Debian, Red Hat, and Ubuntu as the preferred distributions of Linux. However, many other distributions are supported. For more information, see the Technical Support page.

Digium's software, including drivers and application software, is available for download from the Digium download server at:

<http://downloads.digium.com>

For an introduction to Asterisk, Digium's telephony software, and for additional information on its configuration, setup, and features, please refer to:

<http://www.asterisk.org>

For the latest information on setting up and configuring Digium hardware for your Digium hardware product, please refer to the latest product manual which is available from the product-specific documentation page.

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- Development libraries and headers for newt
- GCC and standard software build tools

It is recommended that you use the most recent version of DAHDI, and libpri software for the best results. If you have installed any of these, Digium recommends that you upgrade to the “-current” version of each.

If you are using the 1.4.x series of Asterisk, you will need 1.4.23 or newer.

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Vendor ID which is "d161". The screen output should be the following:

```
0000:01:00.0 0200: ISDN controller: Unkn  
d161:<card identifier>
```

Note: The output from **lspci** may or may not state "ISDN controller" or "device". If it does, this does not indicate a problem.

In the PCI device listing shown above, <card identifier> is populated with one of the identifiers listed in the table below.

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A Digium TE120 Series (TE120P/TE121/TE122) card should be listed. If a matching card identifier is not listed, your machine is not PCI 2.2 (or higher) or PCI Express compatible. Your card will not work with your motherboard.

3. Download the latest version of libpri. Substitute the version number for the X.X in the command line below. libpri is available for download from:

<http://downloads.digium.com/pub/telephony/libpri>

```
# wget http://downloads.digium.com/pub/telephony/libpri/libpri-X.X.current.tar.gz
```

4. Expand the downloaded file, compile its contents, and install the libraries. Substitute the version of libpri for the X.X and

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Note: There is no coorelation between the versioning Asterisk. The libpri 1.4 branch will function with the 1.4 branch.

5. Download the latest DAHDI drivers with tools. DAHDI for download from:

<http://downloads.digium.com/pub/telephony/dahdi-linux-complete>

```
# wget http://downloads.digium.com/pub/telephony/dahdi-linux-complete/dahdi-linux-complete-current.tar.gz
```

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```
# make install  
# make config
```

Note: Executing ‘make config’ will install an init script which will allow you to start and stop DAHDI as a service.

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```
# wget http://downloads.digium.com/pub/telephony/asterisk/asterisk-X.X-current.tar.gz
```

2. Expand the downloaded file, compile its contents, and install the application. Substitute the version of Asterisk for the text X.X.X in the command lines below.

```
# tar -zxvf asterisk-X.X-current.tar.gz
# cd asterisk-X.X.X/
# ./configure
# make menuselect
# make
# make install
```

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If your installation has failed, it may be because you are missing one or more of the build dependencies, the kernel headers, or the development tools. Please contact your reseller where you purchased, or call Digium Technical Support at 1.256.386.4789 for assistance.

4. Complete instructions for installing Asterisk are available at www.asterisk.org.

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finally, data mode. These sample configurations are provided to help you get started in familiarizing yourself with the flexibility of editing the configuration files to meet your specific needs. The list of possible configurations is too expansive to cover in this user manual.

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```
dms100:      Nortel DMS100
4ess:        AT&T 4ESS
5ess:        Lucent 5ESS
euroisdn:    EuroISDN
nil:         Old National ISDN 1
qsig:        Q.SIG
```

Echocancel:

Echo Cancellation is enabled in `chan_dahdi.conf` by preconfiguring a channel variable with a variable called `echocancel` and its value (in milliseconds multiplied by 8); for example:

```
echocancel = yes

channel => 1-23
```

By default, and when setting to "yes," echo cancellation is set to 16 ms (128 taps). Echo cancellation is explicitly disabled by setting `echocancel` to `no`.

```
echocancel = no
```

Digium does not recommend that users set echo cancellation to `yes`.

Users of open source Asterisk-based echo cancelers also have the following options:

```
echocancel = 128 (this sets 128 taps or 16 ms)
```

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$$\text{echocancel} = 1024 - (\text{chip_bits} - 1024) \times \text{taps} \times 0.000001$$

Please note that HPEC consumes extremely high amounts of memory that increase as the number of taps are increased. Audio quality may result from choosing a taps length greater than the sample rate to process the echo in real-time. If audio quality is affected, please contact length or purchase a TE121 or TE122 with Digium's VPMADT032.

Users of Digium's VPMADT032 hardware echo cancellator will not have 128ms of echo cancellation performed at all times unless it is disabled by setting the echocancel variable equal to "no."

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```
switchtype = national  
group = 1  
context = incoming  
channel => 1-23
```

E1 PRI

```
signalling = pri_cpe  
switchtype = euroisdn  
context = incoming  
channel => 1-15,17-31
```

You can also configure a T1 channel bank of phones

```
signalling = fxo_ks  
group = 1  
context = phones  
channel => 1-24
```

E1 channel bank

```
signalling = fxo_ks  
group = 1  
context = phones  
channel => 1-15,17-31
```

Note: More detailed troubleshooting information is provided at www.asterisk.org.

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your interfaces.

The following is a typical setup for a telco in the US:

```
loadzone = us  
defaultzone = us
```

3. Configure the SPAN Map.

You will need to define a span. The SPAN map includes SPAN number, timing, line build out, framing, and coding. Configuration details for each of these items is explained in the next section.

```
span => <Number>,<Timing>,<Line Build  
Out>,<Framing>,<Coding>[,Yellow]
```

Number:

This is the port the T1/E1 is plugged into. The TE120 Series only have one port. If you have a single TE120 Series in the system, you should specify 1 as the span number.

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- 3 - Receives tertiary backup timing from remote end
- 4 - Receives quaternary backup timing from remote end

Only one span can be defined to take timing, and it defines the rest of the card's spans.

Line Build Out:

For most setups the line build out is 0.

- 0: 0 db (CSU) / 0-133 feet (DSX-1)
- 1: 133-266 feet (DSX-1)
- 2: 266-399 feet (DSX-1)
- 3: 399-533 feet (DSX-1)
- 4: 533-655 feet (DSX-1)
- 5: -7.5db (CSU)
- 6: -15db (CSU)
- 7: -22.5db (CSU)

Framing:

T1 utilizes framing set for D4 (SF) or ESF. E1 utilizes

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yellow alarm when no channels are open.

The following is a typical setup for a telco in the US:

```
span => 1,1,0,esf,b8zs
```

In Europe:

```
span => 1,1,0,ccs,hdb3
```

4. Specify the channel definitions. The format is:

```
<device> = <channel list>
```

A list of valid devices are specified in the sample system

The following is a typical setup for a T1 PRI in the US:

```
bchan = 1-23
```

```
dchan = 24
```

5. DAHDI uses modular echo cancellers that are configured in the DAHDI configuration file. The echo cancellers are compiled and installed as part of the DAHDI linux package. You can specify the echo canceller to be used for each channel. The default behavior is for there to be no echo cancellation.

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The following is a typical setup for a T1 PRI in the US based echo cancellation:

```
echocanceller = mg2,1-23
```

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```
/etc/dahdi/system.conf:  
loadzone = us  
defaultzone = us  
span = 1,0,0,esf,b8zs  
fxoks = 1-24  
echocanceller = mg2,1-24
```

```
/etc/asterisk/chan_dahdi.conf:  
group = 1  
echocancel = yes  
context = channelbank  
signalling = fxo_ks  
channel = 1-24
```

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```
/etc/dahdi/system.conf:
loadzone = us
defaultzone = us
span = 1,1,0,esf,b8zs
e&m = 1-24
echocanceller = mg2,1-24

/etc/asterisk/chan_dahdi.conf:
group = 1
echocancel = yes
context = incoming
signalling = feat_d
channel = 1-24
```

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PRI T1:

```
/etc/dahdi/system.conf:
loadzone = us
defaultzone = us
span = 1,1,0,esf,b8zs
bchan = 1-23
dchan = 24
echocanceller = mg2,1-23

/etc/asterisk/chan_dahdi.conf:
group = 1
echocancel = yes
signalling = pri_cpe
switchtype = national
context = incoming
channel = 1-23
```

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```
echocanceller = mg2,1-15,17-31

/etc/asterisk/chan_dahdi.conf:
group = 1
echocancel = yes
signalling = pri_cpe
switchtype = euroisdn
context = incoming
channel = 1-15,17-31
```

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```
span = 1,0,0,esf,b8zs
nethdlc = 1-24
```

Instructions for Cisco HDLC:

- 1. Compile the Linux kernel with Cisco HDLC support.
kernel menuconfig menu may look similar to the follo

Device Drivers --->

[*] Network device support --->

[*] Wan interfaces support --->

<M> Generic HDLC support

<M> Cisco HDLC support

Note: Digium recommends using Linux kernel version 2.4.18 or later.
The HDLC implementation in Linux kernel versions prior to 2.4.18
may not be reliable or function at all.

- 2. Install the newly compiled Linux kernel.

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5. Rebuild DAHDI in order to compile the **sethdlc** utility and execute the following commands from the DAHDI complete source directory:

```
# make sethdlc
# make install
```

6. Load and configure the driver:

```
# modprobe wctel2xp
# dahdi_cfg
```

7. Use **sethdlc** to bring up the interface:

```
# sethdlc hdlc0 cisco
```

8. Assign the interface an address:

```
# ifconfig hdlc0 192.168.0.1 netmask 255.255.255.0
```

The interface may be addressed as any other networking interface (i.e., eth0) in Linux.

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```
# dahdi_cfg -vv
# dmesg
```

```
ACPI: PCI interrupt 0000:01:00.0[A] -> GSI 21 (level
209
PCI Config reg is 02900117
wctel120p: New Reg: fe590000!
Detected REG0: 00000100
Detected REG1: 00007849
Detected REG2: 0000001d
(pre) Reg fc is 50000027
Detected REG0: 0000ffff
(post) Reg fc is 50000024
Detected REG2: 0000ffff
wctel120p: reg is a04c0004
TE120P: FALC version: 00000000
TE120P: Setting up global serial parameters for T1
TE120P: Successfully initialized serial bus for card
Found a Wildcard TE: Wildcard TE120P
```

Figure 8: Example dmesg Output

Note: Output as shown above may vary depending on Series card you use.

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asterisk
asterisk -vvvr

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What do the Status LED colors indicate?

- Green - Card is in-sync with the far end.
- Yellow - Card is synchronizing or is receiving a red alert from the far end. Use a software tool such as dahdi_tool to get a text description of the state of the card.
- Red - Card is not seeing far end, circuit is not up, or card is not working.

I can't receive DID calls even though I have it enabled in extensions.conf.

Your telco might be sending calls with a method you are not expecting.

1. Check the method being used by attempting the following context:

```
_X.,1,NoOp(My DID matches as ${EXTEN})
```

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I have trouble dialing out. It seems that one type of dialing works (local, long distance, international), but another does not.

Check your **pridialplan** variable and verify that you are dialing by the method your telco is expecting.

I am having trouble receiving DID information over E&M.

Try the other types of E&M (featd, featb, etc.) to match the type that the telco is using to stream information.

I am having issues with my PRI. How can I see the messages coming across my D channel?

Enter the following command:

```
*CLI> PRI debug span X
```

where X is the port from which you are connected. This command will show you the PRI messages coming across your D channel.

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2. Stop Asterisk and edit **system.conf** by removing the lines related to your card and replacing them with the following:

```
span => 1,0,0,esf,b8zs  
clear = 1-24
```

Or if you have an E1 span:

```
span => 1,0,0,ccs,hdb3  
clear = 1-31
```

3. Navigate to the tools/ directory in your DAHDI compilation directory and type:

```
# make tests
```

Followed by:

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This runs a pattern looptest for 60 seconds. If you receive a busy signal, it is possible you have a bad card and will need to call Digium Technical Support at 1.256.428.6161

How can I enable more features?

To view all of the options available to add to your dial plan, enter the following commands from within Asterisk:

```
*CLI> core show applications
```

```
*CLI> core show functions
```

Digium also offers services to help configure and add features to your system. Contact Digium Technical Support at 1.256.428.6161 for more information.

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If the X Window System is running, stop the application to avoid
cause a conflict with Asterisk.

- 2. Check to see if your PATA IDE hard drives are running at the correct
levels set. Advance user can perform an **hdparm** on your IDE
interface.



*Use hdparm with caution as the man page states that data
corruption can occur when using incorrect settings. Always
review the man page for hdparm and make sure you understand
the risks before using this tool.*

Check the current mode using this command:
hdparm -vi /dev/[IDE Device]

Use this command to set the drives into UDMA2 mode:
hdparm -d 1 -X udma2 -c 3 /dev/[IDE Device]

If you are still having problems, contact your reseller or the
card was purchased, or Digium Technical Support at 1-800-393-9753.

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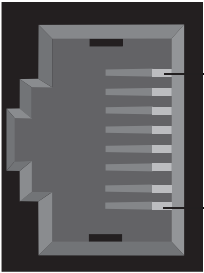
2. Asterisk users mailing list (asterisk.org/lists.digium.com).
3. IRC channel **#asterisk** on (irc.freenode.net).

Subscription Services Program

Digium is dedicated to supporting your Asterisk system by providing technical support through our Subscription Services Program. Through this program, you can be at ease knowing that your business needs will have access to the Asterisk experts. Pricing on Subscription Services can be obtained from your nearest reseller or you may call Digium for a referral to your nearest reseller at +1.256.428.6000 or e-mail sales@digium.com.

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Table A-1: RJ45 Telco Port Connector

| | Pin | Desc |
|---|-----|------|
|  <p>Pin 1</p> <p>Pin 8</p> | 1 | |
| | 2 | |
| | 3 | Not |
| | 4 | |
| | 5 | |
| | 6 | Not |
| | 7 | Not |
| | 8 | Not |

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cards.

Physical (All Cards).

Size: 4.82” × 2.175” × 0.63” (12.2 x 5.5 x 1.6 cm)
PCB size, does not include the PCI bracket
Weight: 2 oz (57g)

Interfaces.

Local Loop Access: E1, T1, J1, PRI; RJ45
(TE120P and TE122) - PCI Bus: 3.3V or 5V bus slot
slot minimum size, 33MHz minimum bus speed, conforming to
PCI 2.2 or greater.
(TE121) - PCI-E X1, compliant with PCI-E X1 1.0 or greater

Environment.

Temperature: 0 to 50° C (32 to 122° F) operation
-20 to 70° C (4 to 158° F) storage
Humidity: 10 to 90% non-condensing

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Table B-2: Maximum Power Consumption

| Model | Power |
|----------------------|----------------------|
| TE120P 3.3V 5V | 1.5 Watts 0.1Watt |
| TE121 3.3V | 2.0 Watts |
| TE121B 3.3V | 3.0 Watts |
| TE122 3.3V 5V | 1.5 Watts 0.1Watt |
| TE122B 3.3V 5V | 2.5 Watts 0.1Watt |

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international communications.

asynchronous

Not synchronized; not timed to an outside clock source. Transmission is controlled by start bits at the beginning and stop bits at the end of each character. Asynchronous communications are often found in dial-up access and remote office applications.

attenuation

The dissipation of a transmitted signal’s power as it travels over a medium.

bandwidth

The capacity to carry traffic. Higher bandwidth indicates the ability to transfer more data in a given time period.

bit

The smallest element of information in a digital system. A bit can be either a zero or a one.

bps *bits per second*

A measurement of transmission speed across a data connection.

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A generic term for an individual data stream. Service providers use multiplexing techniques to transmit multiple channels over a single medium.

Cat5

Category of Performance for wiring and cabling. Cat 5 cabling supports applications up to 100 MHz.

Cat5E

Category of Performance for wiring and cabling. Category 5E wiring supports signal rates up to 100 MHz but adheres to Category 5 specifications.

CLEC *competitive local exchange carrier*

A term for telephone companies established after the Telecommunications Act of 1996 deregulated the LECs. CLECs compete with ILECs to offer local service. See also *LEC* and *ILEC*.

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terminal equipment which is connected to the telecommunication network and which resides within the home or office of the user. This includes telephones, modems, terminals, routers, and teleconferencing boxes.

DAHDI *Digium Asterisk Hardware Device Interface*

A telephony project dedicated to implementing a reasonably affordable computer telephony platform into the world market. It is the collective name for the Digium-provided drivers for Linux-based computer telephony interface products.

DS0 *Digital Signal, Level 0*

A voice grade channel of 64 Kbps. The worldwide standard for digitizing voice conversation using PCM (Pulse Code Modulation).

DS1 *Digital Signal, Level 1*

1.544 Mbps in North America (T1) and Japan (J1) - up to 24 channels (DS0s), 2.048 Mbps in Europe (E1) - up to 32 channels (DS0s). DS1/T1/E1 lines are part of the PSTN.

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E1

The European equivalent of North American T1, transmitting 2.048 Mbps, up to 32 voice channels (DS0s).

E3

The European equivalent of North American T3, transmitting 46.080 Mbps, up to 512 voice channels (DS0s). Equivalent to 16 E1s.

EMI *Electromagnetic Interference*

Unwanted electrical noise present on a power line

full duplex

Data transmission in two directions simultaneously.

FXO *Foreign Exchange Office*

Receives the ringing voltage from an FXS device.

FXS *Foreign Exchange Station*

Initiates and sends ringing voltage.

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A recommendation by the Telecommunication Standardization Group (ITU-T) for an algorithm designed to transmit and receive voice over telephone lines at 6.3 Kbps or 5.3 Kbps.

G.729a

A recommendation by the Telecommunication Standardization Group (ITU-T) for an algorithm designed to transmit and receive voice over telephone lines at 8 Kbps.

H.323

A recommendation by the Telecommunication Standardization Group (ITU-T) for multimedia communications over packet-based networks.

IAX *Inter-Asterisk eXchange*

The native VoIP protocol used by Asterisk. It is an IETF standard that enables VoIP connections between Asterisk servers, and between Asterisk servers and clients that also use the IAX protocol.

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The LECs that were the original carriers in the market prior to the entry of the new entrants have the dominant position in the market and therefore have the dominant position in the market.

interface

A point of contact between two systems, networks, or devices.

ISO *International Standards Organization*

LED *light-emitting diode*

Linux

A robust, feature-packed open source operating system based on the GNU/Linux kernel that remains freely available on the internet. It boasts deep compatibility with hardware and software and offers a wide range of compatibility with hardware and software. Linux is supported exclusively on Linux.

loopback

A state in which the transmit signal is reversed back as the receive signal, typically by a far end network element.

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MUX *multiplexer*

A device which transmits multiple signals over a single communication line or channel. See multiplexing.

PBX *private branch exchange*

A smaller version of a phone company's large central switch. Example: Asterisk.

PCI *peripheral component interconnect*

A standard bus used in most computers to connect peripheral devices.

POP *point of presence*

The physical connection point between a network and a telecommunications network. A POP is usually a network node serving as the connection point between a CO to a network service provider or an interexchange carrier.

POTS *plain old telephone service*

Standard phone service over the public switched telephone network (PSTN). This service provides analog bandwidth of less than 4 kHz.

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world's public circuit-switched telephone networks. Originally, the PSTN was composed of fixed-line analog telephone systems, the PSTN is now digital, and now includes mobile as well as fixed telephones.

QoS *quality of service*

A measure of telephone service, as specified by the Public Utility Commission.

RJ11

A six-pin jack typically used for connecting telephones, modems, and fax machines in residential and business settings to PBX or to a central office telephone CO.

SIP *Session Initiation Protocol*

An IETF standard for setting up sessions between one or more endpoints. SIP is currently the leading signaling protocol for Voice over IP, replacing H.323.

T1

A dedicated digital carrier facility which transmits up to 24 channels (DS0s) and transmits data at 1.544 Mbps. Commonly used to carry traffic to and from private business networks and ISDN.

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into a single high-speed data stream. TDM separates signals by time, interleaving bits one after the other.

telco

A generic name which refers to the telephone companies of the world, including RBOCs, LECs, and PTTs.

tip and ring

The standard termination on the two conductors of a telephone line, named after the physical appearance of the contact areas on the plug.

twisted pair

Two copper wires commonly used for telephony and data communications. The wires are wrapped loosely around each other to minimize radio frequency interference or interference from other wires in the same bundle.

V *volts*

VoIP *Voice over IP*

Technology used for transmitting voice traffic over a data network using the Internet Protocol.

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