

OpenVox Communication Co.Ltd

OpenVox-Best Cost Effective Asterisk Cards

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Test Environments

CentOS-5.6

Kernel version: 2.6.18-238.12.1.el5

DAHDI: dahdi-linux-complete-2.3.0.1+2.3.0

Asterisk: 1.8.0

ss7: chan_ss7-2.1.0

Hardware: IX Series (A400E+DE130E)

Chapter 1 Software Installation

IX series IP-PBX supports combinations analog, BRI, PRI and GSM telephony cards, now let's take the combination of A400E and DE130E for an example to illustrate software installation. And assume that DE130E need to run in SS7 signaling.

1.1 Download

Download chan_ss7 package from:

http://www.netfors.com/chan_ss7_free

Get DAHDI source package from openvox:

http://downloads.openvox.cn/pub/drivers/dahdi-linux-complete/openvox_dahdi-linux-complete-current.tar.gz_

Get Asterisk software package from digium official website:

http://downloads.asterisk.org/pub/telephony/asterisk/releases/asterisk-1.8
.0.tar.gz

Execute the following commands in the directory /usr/src/ in general to download and unzip the three source packages.

- # wget http://www.netfors.com/media/download/chan_ss7-2.1.0.tar.gz
- # wget http://downloads.openvox.cn/pub/drivers/dahdi-linux-complete/openvox_dahdi-linux-complete-current.tar.gz
- # wget http://downloads.asterisk.org/pub/telephony/asterisk/releases/asterisk-1.

8.0.tar.gz

```
# tar -xvzf chan_ss7-2.1.0.tar.gz
```

tar -xvzf openvox_dahdi-linux-complete-xx.tar.gz

tar -xvzf asterisk-1.8.0.tar.gz

1.2 Installation

Make sure your <u>necessary dependencies</u> have all been installed, and then get start to install DAHDI, Asterisk and chan_ss7.

cd /usr/src/dahdi-linux-complete-XX

make

make install

make config



Caution: If there is something wrong after "make", please refer to <u>HERE</u>. In the url link, the moderator introduces you a method how to patch. After patching,

save your changes and exit. Then run "make" again, if successfully, you are going to install Asterisk.

Please operate those commands to install Asterisk.

cd asterisk-1.8.0

./configure

make

make install

make samples



"make samples" will install the standard sample configuration file in the directory /etc/asterisk. As a freshman, you should perform make samples, that is to say,

it is unnecessary to perform make samples every time. Because once performed, it will cover the old sample configuration files you have installed.

After installed dahdi and asterisk, now install chan_ss7 as below:

Go to the directory of chan_ss7 source codes and modify Makefile like this:

#INCLUDE+=-I../source/telephony/zaptel/kernel

#CFLAGS+=-DUSE_ZAPTEL

ASTERISK_PATH=../asterisk-1.8.0/include

INCLUDE+=-I../dahdi-linux-complete-2.3.0.1+2.3.0/include

Execute commands to compile and install chan_ss7

make

make install

cp ss7.conf /etc/asterisk

; copy the ss7.conf file to /etc/asterisk

cp chan_ss7.so /usr/lib/asterisk/modules

; copy the chan_ss7.so to /usr/lib/asterisk/modules

1.3 Configuration

1. Driver loading

After compiling and installing DAHDI and Asterisk, please load the driver by running:

- # modprobe dahdi
- # modprobe opvx115
- # modprobe wctdm
- # dahdi_genconf



After running, there is not any indication information displayed if loaded normally and successfully. "opvx115" and "wctdm" are the driver module name of DE130E and A400E.

If there is any error, please trace the cause. Until all errors are clear up, you could execute "dahdi_genconf" again, and then go to the next step. By running "dahdi_genconf", it will generate /etc/dahdi/system.conf and etc/asterisk/dahdi-channels.conf automatically. Checking whether the generated files information agrees with your hardware setup, if not, you

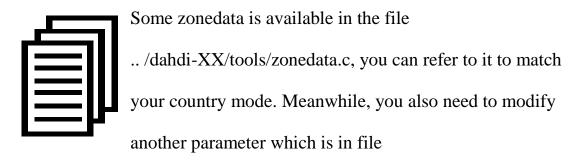
should modify to your specific requirements. Do not forget to confirm dahdi-channels.conf is included in chan dahdi.conf, if not, run command:

echo "#include dahdi-channels.conf" >> /etc/asterisk/chan_dahdi.conf

FXO ports use FXS signaling, while FXS ports adopt FXO signaling. A part of system.conf which is one of the basic channel configuration files is displayed.

```
# Span 2: WCTDM/4 "Wildcard TDM400P REV E/F Board 5" (MASTER)
fxoks=1
fxoks=2
fxsks=3
fxsks=4
# Global data
loadzone = cn
defaultzone = cn
;Modify these two parameters to peer your country or zone
```

Figure 6 A part of system.conf



/etc/asterisk/indications.conf.

country=cn

2. Edit files for ss7

a. Edit /etc/dahdi/system.conf like:

```
vim /etc/dahdi/system.conf
This file is parsed by the Dahdi Configurator, dahdi cfg
# Span 1: TE4/0/1 "T4XXP (PCI) Card 0 Span 1" HDB3/CCS/CRC4 RED
Span=1,1,0,ccs,hdb3,crc4
# termtype: te
bchan=1-31
#dchan=16
```

b. edit /etc/asterisk/ss7.conf

```
[linkset-siuc]
enabled => yes
enable st => no
use connect => no
hunting policy => even mru
context => ss7
language => da
t35 => 15000, timeout
subservice => auto
;signallingtype=>ISUP(TUP)
variant => CHINA
[link-l1]
linkset => siuc
channels => 1-15,17-31
schannel => 16
firstcic => 0
;sls => 1
;sltm => no
enabled => yes
echocancel => no
;echocan train => 350
;echocan taps => 128
[host-zmdvoip]
                           zmdvoip is your asterisk server name
enabled => yes
opc => 0x10ff48
dpc => siuc:0x10fff6
links => 11:1
```



Some parameters in this file need to be changed according to your situation. Please replace "zmdvoip" by your asterisk server name. If "opc" is 0x10fff6 and "dpc" is 0x10ff48 carrier gives you, you must set

"opc => 0x10ff48 dpc => siuc:0x10fff6" like the above figure.

c. Edit /etc/asterisk/dahdi-channels.conf

Please comment out the settings for DE130E, otherwise it will be contradictory with ss7

```
;Span 1: TE4/0/1 "T4XXP (PCI) Card 0 Span 1" HDB3/CCS/CRC4 RED
;group=0,11
;context=from-pstn
;switchtype = euroisdn
;signalling = pri cpe
;channel => 1-15,17-31
;context = default
; group = 63
```

After these tasks finished, please execute the following command:

```
# service dahdi restart
# dahdi_cfg -vvvvvv
```

The command is used for reading and loading parameters in the configuration file system.conf and writing to the hardware. A part of outputs are showed in the following figure.

```
DAHDI Version: 2.3.0
Echo Canceller(s):
Configuration
Channel map:
Channel 01: FXO Kewlstart (Default) (Echo Canceler: none) (Slaves: 01)
Channel 02: FXO Kewlstart (Default) (Echo Canceler: none) (Slaves: 02)
Channel 03: FXS Kewlstart (Default) (Echo Canceler: none) (Slaves: 03)
Channel 04: FXS Kewlstart (Default) (Echo Canceler: none) (Slaves: 04)
4 channels to configure.
Setting echocan for channel 1 to none
Setting echocan for channel 2 to none
Setting echocan for channel 3 to none
Setting echocan for channel 4 to none
```

Figure 7 Channel map

3. Asterisk initiation

asterisk -vvvvvvvgc

If Asterisk is already activate, run "asterisk -r" instead. In the CLI, please run the following command:

localhost*CLI> ss7 link status

linkset siuc, link 11/16 INSERVICE, sls 0, total: 8684336, 8684368

localhost*CLI> ss7 linestat

```
Linkset: siuc
```

- CIC 0 Idle
- CIC 1 Idle
- CIC 2 Idle
- CIC 3 Idle
- CIC 4 Idle
- CIC 5 Idle
- CIC 6 Idle
- CIC 7 Idle
- CIC 8 Idle
- CIC 9 Idle
- CIC 10 Idle
- CIC 11 Idle
- CIC 12 Idle
- CIC 13 Idle
- CIC 14 Idle
- CIC 16 Idle
- CIC 17 Idle
- CIC 18 Idle
- CIC 19 Idle
- CIC 20 Idle
- CIC 21 Idle
- CIC 22 Idle
- CIC 23 Idle
- CIC 24 Idle
- CIC 25 Idle
- CIC 26 Idle

CIC 27 Idle CIC 28 Idle CIC 29 Idle CIC 30 Idle

localhost*CLI> dahdi show channels

Chan Extension	Context	Language	MOH Interpret
Pseudo	default		default
1	from-internal		default
2	from-internal		default
3	from-pstn		default
4	from-pstn		default

Figure 9 channels show

If DAHDI channels and ss7 are found and up, it means they have been loaded into Asterisk successfully. The last thing is to edit your extension (softphone or hard phone) and dialplan by your requirements. After right dialplan, I will say "congratulations to you!"

After saving your dialplan, please run "asterisk –r", then execute "reload" in the CLI. Next you are able to make calls.

Chapter 2 Reference

www.openvox.cn

www.digium.com

www.asterisk.org

www.voip-info.org

www.asteriskguru.com

Appendix

```
# yum install bison
# yum install bison-devel
# yum install ncurses
# yum install ncurses-devel
# yum install zlib
# yum install zlib-devel
# yum install openssl
# yum install openssl-devel
# yum install gnutls-devel
# yum install gcc
# yum install gcc-c++
# yum install libxml2
```

Tips

Any questions during installation please consult in our forum or look up for answers from the following websites:

Forum

wiki