Raspberry Pi Packet TNC, APRS Digipeater, IGate

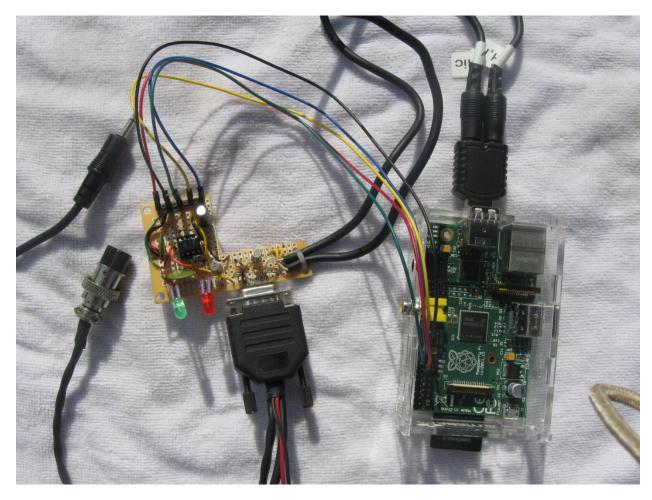
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In the early days of Amateur Packet Radio, it was necessary to use a "Terminal Node Controller" (TNC) with specialized hardware. Those days are gone. You can now get better results at lower cost by connecting your radio to the "soundcard" interface of a computer and running free software.

The Raspberry Pi (RPi) is a good platform for running a software TNC, APRS digipeater, and IGate. Why use a larger computer and waste hundreds of watts of power? All you need to add is a USB Audio Adapter (\$8 or less) and a simple PTT circuit to activate the transmitter.

This document is a Quick Start guide for running Dire Wolf on the Raspberry Pi and describes special considerations where it may differ from other Linux systems.

After completing the steps here, refer to the User Guide for more details on the Linux version.



Left: Connections to speaker and Microphone jack of transceiver.

Middle: Interface circuit with a timer to limit transmission time.

It uses the standard 9 pin connector found on most TNCs and trackers so the same radio-specific cables can be used.

The two LEDs are for Data Carrier Detect (DCD) and the Push to Talk (PTT) signals.

The shape of the board was not intentional. It was just a scrap piece of perfboard left over from another project.

Right: Raspberry Pi.

At the top is a USB audio adapter. The current software version can handle 3 audio interfaces at the same time.

Lower left are GPIO connections for PTT and the DCD LED.

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1 Install Appropriate Operating System

Select a version of Linux that has hardware floating point support. Dire Wolf makes extensive use of floating point calculations. Trying to use the slower software floating point will probably result in disappointment.

The Raspbian "wheezy" distribution from http://www.raspberrypi.org/downloads is known to work with the instructions here. I haven't tried the others and don't know how they might differ.

The Raspian operating system distribution comes with the **gcc** compiler and most required libraries preinstalled. If you use a different operating system version, you might need to install a suitable compiler and/or additional libraries.

1.1 If operating system is already installed.

If you already have a different operating system version, verify that gcc is configured to generate hardware floating point code. Enter the "gcc –v" command and observe the result. Make sure that "--with-fpu=vfp --with-float=hard" appears in the configuration.

```
pi@raspberrypi:~$ gcc -v
Using built-in specs.
COLLECT GCC=qcc
COLLECT LTO WRAPPER=/usr/lib/gcc/arm-linux-gnueabihf/4.6/lto-wrapper
Target: arm-linux-gnueabihf
Configured with: ../src/configure -v --with-pkgversion='Debian 4.6.3-12+rpi1' -
-with-bugurl=file:///usr/share/doc/gcc-4.6/README.Bugs --enable-
languages=c,c++,fortran,objc,obj-c++ --prefix=/usr --program-suffix=-4.6 --
enable-shared --enable-linker-build-id --with-system-zlib --libexecdir=/usr/lib
--without-included-gettext --enable-threads=posix --with-gxx-include-
dir=/usr/include/c++/4.6 --libdir=/usr/lib --enable-nls --with-sysroot=/ --
enable-clocale=gnu --enable-libstdcxx-debug --enable-libstdcxx-time=yes --
enable-gnu-unique-object --enable-plugin --enable-objc-gc --disable-sjlj-
exceptions --with-arch=armv6 --with-fpu=vfp --with-float=hard --enable-
checking=release --build=arm-linux-gnueabihf --host=arm-linux-gnueabihf --
target=arm-linux-gnueabihf
Thread model: posix
gcc version 4.6.3 (Debian 4.6.3-12+rpi1)
```

1.2 New System Setup & Configuration

If you are starting with a brand new RPi, the lessons here: http://learn.adafruit.com/ are easy to follow and will get you running quickly.

- https://learn.adafruit.com/adafruit-raspberry-pi-lesson-1-preparing-and-sd-card-for-your-raspberry-pi
- https://learn.adafruit.com/adafruits-raspberry-pi-lesson-2-first-time-configuration
- https://learn.adafruit.com/adafruits-raspberry-pi-lesson-3-network-setup
- https://learn.adafruit.com/adafruits-raspberry-pi-lesson-4-gpio-setup
- https://learn.adafruit.com/adafruits-raspberry-pi-lesson-6-using-ssh
- https://learn.adafruit.com/adafruit-raspberry-pi-lesson-7-remote-control-with-vnc

The last two are optional, depending on your situation. In my case, I have a headless (no monitor, keyboard) Raspberry Pi in the radio "shack" which is accessed from other computers various places over the network. The VNC software allows me to have a virtual desktop displayed on other computers. If you are using a monitor / keyboard / mouse connected to the RPi, you won't need to configure login by ssh and VNC.

The audio system on the Raspberry Pi has a history of many problems. Every time I did a software update, the behavior changed. Before October 13, 2013, the only way I could get it to work was with pulseaudio. After a most recent software update and firmware upgrade, pulseaudio stopped working (http://elinux.org/R-Pi_Troubleshooting#Removal_of_installed_pulseaudio) but the preferred method, which never worked properly before, is now fine.

You might be dooming your efforts to failure if you skip this software & firmware update step.

```
sudo apt-get update
sudo apt-get dist-upgrade
sudo rpi-update
sudo reboot
```

Check the firmware version with this command. I'm getting good results with the Oct 12 2013 version and later. **Earlier versions were not satisfactory.**

```
/opt/vc/bin/vcgencmd version
```

Pulseaudio was never right and it was even worse the last time I struggled with it. I believe it is no longer included in the current version of Raspbian. Remove it just to be safe.

```
sudo apt-get remove --purge pulseaudio
sudo apt-get autoremove
rm -rf /home/pi/.asoundrc /home/pi/.pulse
```

At this point reboot to make sure you are running the latest versions.

```
sudo reboot
```

1.3 Optionally enable IPv6.

You can enable IPv6 immediately by typing: "sudo modprobe ipv6".

To start it up automatically after each reboot, edit /etc/modules and add a new line at the end containing "ipv6" (without the quotes).

1.4 Make a backup of your SD card (optional)

After going through all of these steps, you might want to make a backup so you can get back to this point quickly if the memory card gets trashed. Here's how: https://www.raspberrypi.org/forums/viewtopic.php?p=239331

2 Install Dire Wolf

2.1 Install sound library.

First, install the "libasound2-dev" package with the following command:

```
sudo apt-get install libasound2-dev
```

Failure to install libasound2-dev step will result in a compile error resembling "audio.c:...: fatal error: alsa/asoundlib.h: No such file or directory"

2.2 Download Dire Wolf source code

Download the source version from http://home.comcast.net/~wb2osz/site/ and copy it to your Raspberry Pi. /home/pi is used as example here. If you put it somewhere else, you will need to make suitable adjustments to the commands shown.

The exact zip file name and unpacked path might vary if you are using a beta test or development version.

```
cd /home/pi
unzip direwolf-1.2-src.zip
cd direwolf-1.2
```

Optionally obtain the most recent updates for the APRS TO-CALL version numbers and the new symbol codes.

```
make tocalls-symbols
```

2.3 Compile and Install

Compile an install the application.

```
make
sudo make install
make install-rpi
```

You should now have files in these locations, under /usr/local, owned by root.

/usr/local/bin/direwolf	The application.
/usr/local/bin/decode_aprs	Utility to interpret "raw" data you might find on
	http://aprs.fi or http://findu.com
/usr/local/bin/tt2text	Utilities related to APRStt gateway, UTM
/usr/local/bin/text2tt	coordinates, log file to GPX conversion, and test

/usr/local/bin/ll2utm	packet generation.
/usr/local/bin/utm2ll	
/usr/local/bin/log2gpx	
/usr/local/bin/gen_packets	
/usr/share/applications/direwolf.desktop	Application definition with icon, command to
	execute, etc.
/usr/local/share/direwolf/tocalls.txt	Mapping from destination address to system type.
	Search order for tocalls.txt is first the current
	working directory and then /usr/share/direwolf.
/usr/local/share/direwolf/symbolsX.txt	Descriptions and codes for APRS symbols.
/usr/local/share/direwolf/symbols-new.txt	
/usr/share/direwolf/dw-icon.png	Icon for the desktop.
/usr/local/share/doc/direwolf/*	Various documentation.
/usr/local/man/man1/*	

You should also have these files, under /home/pi.

/home/pi/Desktop/direwolf.desktop	Symbolic link to
	/usr/share/applications/direwolf.desktop. This
	causes an icon to be displayed on the desktop.
/home/pi/dw-start.sh	Script to start Dire Wolf if it is not running already.

2.4 Get initial configuration file.

If this is the first time, you are installing Dire Wolf perform this step:

```
make install-conf
```

When upgrading from an earlier version, you will probably want to skip this step because it will wipe out your earlier configuration file.

This step should have copied the initial configuration file to the home directory, /home/pi.

/home/pi/direwolf.conf	Configuration file.
	Search order is current working directory then the
	user's home directory.

2.5 First Test

Go to your home directory and try to run "direwolf."

cd /home/pi
direwolf

You should see something like this, because we have not yet configured it for using an audio device.

Dire Wolf version ...
Audio device for both receive and transmit: default
Could not open audio device default for input
No such file or directory
Pointless to continue without audio device.

We will perform the necessary configuration in a later step.

3 Interface for radio

APRS, or other packet radio, operation requires connections between your transceiver and computer.

(a) Received audio from receiver.

This can be very simple. All you need is a cable from the speaker of your radio to the computer's microphone or line in. Some people like to use audio isolation transformers but I never found this to be necessary.

(b) Transmit audio to transmitter.

Audio output from the computer goes to the microphone input of your transceiver. A direct connection might be acceptable in some cases but typically you would want a trim pot to decrease the signal level and a capacitor (perhaps around $0.1~\mu F$) to block DC.

(c) Push to Talk (PTT) signal to activate transmitter.

Traditionally, this has often been with one of the control lines of an RS-232 serial port. It would be possible to use a USB-to-serial adapter but there is a better way. The Raspberry Pi has a bunch of general purpose input output (GPIO) pins.

If you use a VOX circuit, to turn on transmitter automatically when audio is present, no PTT connection would be necessary. You can also ignore the warning message about PTT not being configured.

3.1 Audio input / output

The Raspberry Pi has built-in audio output but no audio input.

You can get a list of audio output devices with the "aplay -I" (that's lower case L option) command.

```
pi@raspberrypi:~$ aplay -1

**** List of PLAYBACK Hardware Devices ****
card 0: ALSA [bcm2835 ALSA], device 0: bcm2835 ALSA [bcm2835 ALSA]
  Subdevices: 8/8
  Subdevice #0: subdevice #0
  Subdevice #1: subdevice #1
  Subdevice #2: subdevice #2
  Subdevice #3: subdevice #3
  Subdevice #4: subdevice #4
  Subdevice #5: subdevice #4
  Subdevice #5: subdevice #6
  Subdevice #7: subdevice #7

card 0: ALSA [bcm2835 ALSA], device 1: bcm2835 ALSA [bcm2835 IEC958/HDMI]
  Subdevices: 1/1
  Subdevice #0: subdevice #0
```

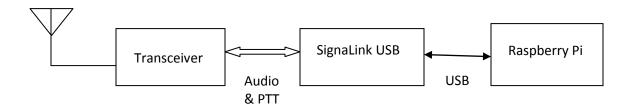
You can get a list of audio input devices with the "arecord -I" (again lower case L option) command:

```
pi@raspberrypi:~$ arecord -1
**** List of CAPTURE Hardware Devices ****
```

There aren't any! It will be necessary to add some sort of sound input device.

3.1.1 Signalink USB

The **Tigertronics SignaLink USB** is a pricey solution but easy because it contains all of the interface circuitry. Be sure to turn the "DLY" delay control completely counter-clockwise.



3.1.2 Cheap USB audio adapters

A cheap USB audio adapter can also be used but the drivers and firmware can be little problematic.

I'm currently using this Syba adapter http://www.amazon.com/gp/product/B001MSS6CS successfully with the Oct 12 2013 version firmware or later. The sticker on the package indicated it has a CMedia HS100 chip (see http://www.cmedia.com.tw/ApplicationIndex/C1Serno-1/C2Serno-4.html) but "lsusb" reports:

Bus 001 Device 004: ID 0d8c:0139 C-Media Electronics, Inc.

The C-Media product id 0139 indicates it's the CM108AH chip. The HS100 has a product id of 013C.

This one looks interesting because it has a short cable which provides more flexibility in the physical arrangement of the various components. http://www.adafruit.com/product/1475

"Isusb" reports this:

Bus 001 Device 005: ID 0d8c:000c C-Media Electronics, Inc. Audio Adapter

This appears to be the CM108 or CM109 chip. According to the CM108 datasheet, the product id can vary from 0008 to 000f and is programmable by the MSEL and MODE pins. The CM109 datasheet says the same thing so I don't know how you can tell them apart.

It works fine on Windows 7 and Linux on a PC but I have not yet been able to get audio input working properly on the RPi even with the latest (December 2014) firmware. I'd like to hear from anyone that is successful.

There is a report of this one spewing out a lot of RFI at 144 MHz so you might want to avoid it even after a properly working driver is available.

3.1.3 Avoid USB hub for audio adapter

Avoid putting a USB hub between the Raspberry Pi and the USB Audio adapter. I've received reports that exactly the same configuration works fine with the direct connection and a significant number of audio samples are lost when going through a hub.

3.2 Determine "card" number for audio device

After plugging in a suitable USB audio interface, you should see something like this, in response to the "aplay –I" and "arecord –I" commands:

```
pi@raspberrypi:~$ aplay -1
      **** List of PLAYBACK Hardware Devices ****
      card 0: ALSA [bcm2835 ALSA], device 0: bcm2835 ALSA [bcm2835 ALSA]
        Subdevices: 8/8
        Subdevice #0: subdevice #0
        Subdevice #1: subdevice #1
        Subdevice #2: subdevice #2
        Subdevice #3: subdevice #3
        Subdevice #4: subdevice #4
        Subdevice #5: subdevice #5
        Subdevice #6: subdevice #6
        Subdevice #7: subdevice #7
      card 0: ALSA [bcm2835 ALSA], device 1: bcm2835 ALSA [bcm2835 IEC958/HDMI]
        Subdevices: 1/1
        Subdevice #0: subdevice #0
(3)
      card 1: CODEC [USB Audio CODEC], device 0: USB Audio [USB Audio]
        Subdevices: 1/1
        Subdevice #0: subdevice #0
      pi@raspberrypi:~$ arecord -1
      **** List of CAPTURE Hardware Devices ****
F
      card 1: CODEC [USB Audio CODEC], device 0: USB Audio [USB Audio]
        Subdevices: 1/1
        Subdevice #0: subdevice #0
```

The built-in (output only) interface will normally be card 0 unless you did something to change the order. The additional USB audio adapter will typically be card 1, as seen in the example above. If you

see a different number, you will have to modify a later step when we put the card number in the configuration file.

3.3 Transmitter PTT

If you want to transmit, some method is needed to activate the transmitter push-to-talk (PTT) function. Traditionally, this has been done with the RTS or DTR signal from a serial port.

Possible alternatives include:

• Reconfigure the built in serial port to have the RTS control.

The Raspberry Pi has a built in serial port which is normally configured as a console. It is possible to reconfigure this to be available for application use. In the default configuration, it does not have the usual RTS/CTS flow control signals. As described here, http://elinux.org/RPi_Serial_Connection#Handshaking_lines, it is possible to reconfigure some of the GPIO lines for the RTS/CTS signals. I haven't tried this.

• Use a USB to RS-232 converter cable to supply a full RS-232 port. You will need to put something like this in the configuration file.

PTT /dev/ttyUSB0 RTS

VOX operation where transmitter is activated by transmit audio signal. The SignaLink USB uses
this technique. (Homebrew circuit example: https://sites.google.com/site/kh6tyinterface/)

I do not recommend using the VOX built into transceivers because they are designed for voice and keep the transmitter on much too long after the audio has stopped. The User Guide has a detailed explanation.

 Use one of the General Purpose I/O (GPIO) pins. I'm currently using this along with the cheap USB audio adapter.

CAUTION! The general purpose input output (GPIO) pins are connected directly to the CPU chip. There is no buffering or other protection. The interface uses 3.3 volts and will not tolerate 5 volt signals. Static discharge, from careless handing, could destroy your Raspberry Pi.

There are many GPIO pins. How would you choose an appropriate one? These are my suggestions for the best choices. The others have special functions such as UART, SPI, PWM, or I²C. A few others changed positions between board revisions 1 & 2 which could cause confusion.

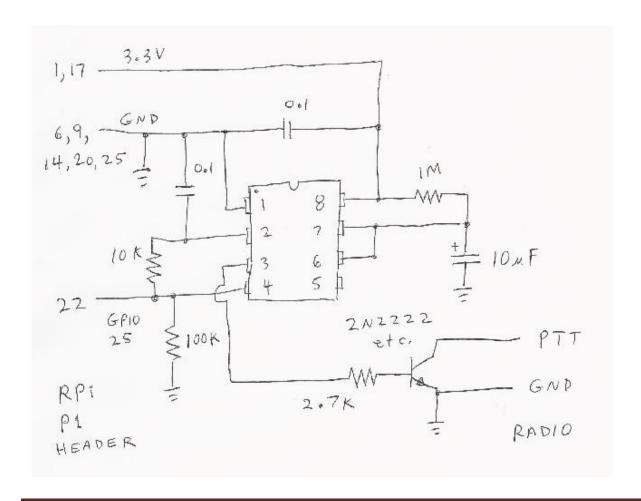
- P1-11 GPIO 17
- P1-15 GPIO 22
- P1-16 GPIO 23
- P1-18 GPIO 24

P1-22 GPIO 25

The A+, B+, and model 2 have a larger connector with additional GPIO pins. Any of them would be suitable:

- P1-29 GPIO 5
- P1-31 GPIO 6
- P1-32 GPIO 12
- P1-33 GPIO 13
- P1-35 GPIO 19
- P1-36 GPIO 16
- P1-37 GPIO 26
- P1-38 GPIO 20
- P1-40 GPIO 21

Here is a recommended circuit using a CMOS 555 timer (LMC555, TLC555, ICM7555, etc.) to limit transmissions to about 10 seconds. Don't try using the original 555 because it needs a minimum of 4.5 volts and we have only 3.3 here. The time can be increased by making the 10 μ F capacitor larger. It's roughly 1 second for each μ F.



You could get by with only a resistor and transistor but a software failure could cause the transmitter to be stuck on, jamming the radio channel, annoying other people, and possibly damaging the transmitter from overheating.

Finally, put a command like this in the configuration file with the actual I/O signal being used.

PTT GPIO 25

4 Configure for Radio Interface

We need to make a couple simple changes to the configuration file. Change to your home directory.

```
cd /home/pi
```

You should find a file called "direwolf.conf." Modify it with your favorite text editor, as explained in the following steps.

4.1 Audio Interface Device

We previously determined the "card" number for the desired audio interface. Typically this will be 1.

Look in the "direwolf.conf" file for a line like this:

```
# ADEVICE plughw:1,0
```

Remove the "#" from the beginning of the line. If the card number, found in the earlier step, is not 1, use that number instead.

4.2 PTT Method

Look in "direwolf.conf" for a line like this:

```
# PTT GPIO 25
```

Assuming you are using GPIO 25 for the PTT control, simply remove the "#" from the beginning of the line. If using some other PTT method, refer to the User Guide for a discussion of other possibilities.

4.3 Optional DCD Indicator

If you would like a data carrier detect (DCD) LED to light up when it looks like a signal is present, add a line like this:

```
DCD GPIO -24
```

This means drive GPIO 24 to ground for signal present. Connect an LED like this:

```
Pin 18 (GPIO 24) ---- (cathode) LED (anode) ---- 270 ohm resistor ---- 3.3 volts
```

5 Run Dire Wolf application

Start up the application with the desktop icon or from the command line:

direwolf

Typical results will look something like this:

```
_ 0
                                                                                           23
pi@raspberrypi: ~
pi@raspberrypi ~ $ direwolf
Dire Wolf version 1.1
Audio device for both receive and transmit: plughw:1,0
Channel 0: 1200 baud, AFSK 1200 & 2200 Hz, F, 44100 sample rate.
Use -p command line option to enable KISS pseudo terminal.
Ready to accept KISS client application on port 8001 ...
Ready to accept AGW client application 0 on port 8000 ...
Now connected to IGate server noam.aprs2.net (2607:fc18:0:3::114)
Check server status here http://[2607:fc18:0:3::114]:14501
[ig] # aprsc 2.0.14-g28c5a6a<0x0d><0x0a>
[ig] # logresp WB2OSZ-5 verified, server T2CMH<0x0d><0x0a>
[OL] WB2OSZ-5>APDW11,WIDE1-1,WIDE2-1:!4237.14NS07120.83W#PHG7140Raspberry Pi digpeater
N5PZ audio level = 5 [NONE]
[0] N5PZ>T2SW9T,WIDE1-1,WIDE2-1: `c-9n5$>/'"4R}|!3&9'?|!wRB!|3
MIC-E, normal car (side view), Byonics TinyTrack3, In Service
N 42 37.9454, W 071 17.2936, 25 MPH, course 108, alt 194 ft
Seq=18, A1=479, A2=576
[OH] N5PZ>T2SW9T,WB2OSZ-5*,WIDE2-1:`c-9n5$>/'"4R}|!3&9'?|!wRB!|3
Digipeater W1MHL audio level = 9 [NONE]
[0] N2PGD-14>APX205,N1HRK-15,WIDE1,W1MHL*,WIDE2:=4149.61N/07123.98W#PHG2200XASTIR-Linux<0x0d
Position, DIGI (white center), Xastir, 4 W height=40 OdBi omni
N 41 49.6100, W 071 23.9800
XASTIR-Linux
Digipeater WIDE1-1 audio level = 7
                                    [NONE]
[0] N5PZ>T2SW9T,WIDE1-1*,WIDE2-1:`c-9n5$>/'"4R}|!3&9'?|!wRB!|3
MIC-E, normal car (side view), Byonics TinyTrack3, In Service
N 42 37.9454, W 071 17.2936, 25 MPH, course 108, alt 194 ft
```

Let's examine each section.

On the first line, we have the application version.

Next we have confirmation of the audio device being used.

The modem is configured for standard 1200 baud operation.

```
Dire Wolf version 1.2
Audio device for both receive and transmit: plughw:1,0
Channel 0: 1200 baud, AFSK 1200 & 2200 Hz, F, 44100 sample rate.
```

This is a summary of the interfaces available for client applications.

```
Use -p command line option to enable KISS pseudo terminal. Ready to accept KISS client application on port 8001 ... Ready to accept AGW client application 0 on port 8000 ...
```

This shows that the station is operating as an Internet Gateway and has successfully connected to a server. A URL is provided to check on the server status. "[ig]" at the beginning of the line indicates a response from the IGate server.

```
Now connected to IGate server noam.aprs2.net (2607:fc18:0:3::114) Check server status here http://[2607:fc18:0:3::114]:14501

[ig] # aprsc 2.0.14-g28c5a6a<0x0d><0x0a>
[ig] # logresp WB2OSZ-5 verified, server T2CMH<0x0d><0x0a>
```

This is a beacon, identifying the station. "[OL]" means it is transmitting on radio channel 0 at low priority. Notice that transmitted lines are displayed in magenta.

```
[OL] WB2OSZ-5>APDW11,WIDE1-1,WIDE2-1:!4237.14NS07120.83W#
PHG7140Raspberry Pi digpeater
```

Here we are receiving a nearby station directly. "[0]" means it was heard on radio channel 0. Received signals are in green.

```
N5PZ audio level = 5 [NONE]
[0] N5PZ>T2SW9T,WIDE1-1,WIDE2-1:`c-9n5$>/'"4R}|!3&9'?|!wRB!|3
```

What to all those strange characters mean? The next few lines contain the decoded information in human readable form. In this example, the last line is telemetry data.

```
MIC-E, normal car (side view), Byonics TinyTrack3, In Service N 42 37.9454, W 071 17.2936, 25 MPH, course 108, alt 194 ft Seq=18, A1=479, A2=576
```

This packet is a candidate for digipeating so we retransmit it. "[0H]" means it is being transmitted on channel 0. Digipeated packets go out at a higher priority. Notice how "WIDE1-1" was replaced by "WB2OSZ-5*" so the actual path taken can be determined.

```
[OH] N5PZ>T2SW9T,WB2OSZ-5*,WIDE2-1:`c-9n5$>/'"4R}|!3&9'?|!wRB!|3
```

Now we hear another station. But this time it is not direct. We are actually hearing a digipeater. It is not a candidate for digipeating because the via path has been all used up. In this case, the last line is the comment.

```
Digipeater W1MHL audio level = 9 [NONE]
[0] N2PGD-14>APX205,N1HRK-15,WIDE1,W1MHL*,WIDE2:=4149.61N/07123.98W#
PHG2200XASTIR-Linux<0x0d>
Position, DIGI (white center), Xastir, 4 W height=40 OdBi omni
N 41 49.6100, W 071 23.9800
XASTIR-Linux
```

Now we hear the first station again, but from a digipeater, not directly. The digipeater did not insert its own call so we don't know who retransmitted the packet.

```
Digipeater WIDE1-1 audio level = 7 [NONE]
[0] N5PZ>T2SW9T,WIDE1-1*,WIDE2-1:`c-9n5$>/'"4R}|!3&9'?|!wRB!|3
MIC-E, normal car (side view), Byonics TinyTrack3, In Service
N 42 37.9454, W 071 17.2936, 25 MPH, course 108, alt 194 ft
Seq=18, A1=479, A2=576
```

This is also a candidate for digipeating because "WIDE2-1" has not been used up yet. However, we transmitted this same packet within the past 30 seconds so the duplicate is dropped.

```
Digipeater: Drop redundant packet.
```

That's all you will see. No flashy graphics. No maps. No user interaction. But loads of valuable information for monitoring activity and troubleshooting problems.

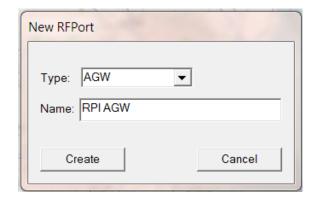
Dire Wolf is just a replacement for a TNC. To do more interesting things, you will need an application such as APRSISCE/32, UI-View32, Xastir, APRS-TW, YAAC, SARTrack, RMS Express, etc.

5.1 Use with client applications

Client applications can run on different computers and communicate with Dire Wolf over your local network. For example, you might have a Raspberry Pi in your "shack," in a cold damp basement, connected to your radio equipment. You might want to use a client application (such as APRSIS32, YAAC, or Xastir) on a laptop in a more comfortable location, perhaps next to the pool or fireplace. Here is an example of how you could configure APRSISCE/32 in this situation.

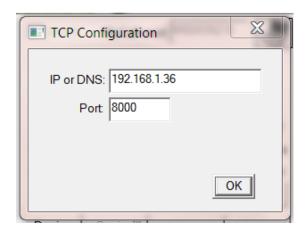
From the Configure menu, pick Ports → New Port...

Choose type of AGW and give it a meaningful name.



Click on Create and pick port type of TCP/IP.

Enter the IP address of the Raspberry Pi, and the default port of 8000.



6 Other Common Configuration Changes

6.1 Automatic Startup

You probably want your TNC / application server / digipeater to start up automatically after a power interruption.

If you followed the installation steps above, you should have a file named dw-start.sh in your home directory. Run the "crontab –e" command and add a line like this:

```
* * * * * /home/pi/dw-start.sh >/dev/null 2>&1
```

This script will run once per minute. Dire Wolf is started automatically if not running already. If it crashes, or is terminated for any other reason, it will be restarted. A log of restarts can be found in /tmp/dw-start.log.

6.2 Digipeater Operation

Edit the /home/pi/direwolf.conf file and look for a section like this:

```
# Station identifier for this channel.
# Multiple channels can have the same or different names.
#
# Naturally it must be up to letters and digits with an optional ssid.
# The APRS specification requires that it be upper case.
#
#
# Example (don't use this unless you are me): MYCALL WB2OSZ-5
#
```

MYCALL NOCALL

Change "NOCALL" to your ham radio call and optional SSID.

Next, look for a section like this:

Remove the "#" character from the beginning of the last line shown above. It is necessary to stop and restart the application to notice configuration file changes.

Here is an example of what you should see:

```
KB1UTS audio level = 5 [NONE]
[0] KB1UTS>APWW10,WIDE1-1:@021038h4235.16N/07121.53W-
Position with time, House QTH (VHF), APRSISCE win32 version
N 42 35.1600, W 071 21.5300
[0H] KB1UTS>APWW10,WB2OSZ-5*:@021038h4235.16N/07121.53W-
```

Dark green: Information about the station we heard. Either the originating station or a digipeater.

Green: Raw received data. Notice that the digipeater field contains "WIDE1-1."

Blue: An explanation for troubleshooting. The destination (APWW10) is used to determine the type of system generating the signal.

Magenta: This is the re-transmitted packet. Notice that the digipeater field now contains "WB2OSZ-5*." The "*" indicates that it has been used up and won't be digipeated again.

6.3 Enable Beaconing

Be sure that MYCALL has been set as in previous step.

Look for a section like this in direwolf.conf file.

Remove the "#" comment character from the beginning of the "PBEACON" lines. Make necessary adjustments to latitude, longitude, comment, etc. These options, and more, are described in the User Guide.

Note that each position beacon command must be on a single line. Multiple lines are shown above due to page width limitations.

6.4 Internet Gateway (IGate)

Dire Wolf can serve as a gateway between the radio network and servers on the Internet. This allows information to be retrieved from locations such as http://findu.com. Information can optionally be relayed from the servers, through your station, and on to the radio.

First you need to specify the name of a Tier 2 server. The current preferred way is to use one of these regional rotate addresses:

noam.aprs2.net
 soam.aprs2.net
 euro.aprs2.net
 for North America
 for South America
 for Europe and Africa

asia.aprs2.net - for Asiaaunz.aprs2.net - for Oceania

Each name has multiple addresses to achieve load balancing and resiliency. Visit http://aprs2.net/ for the most recent information. You also need to specify your login name and passcode. For example:

```
IGSERVER noam.aprs2.net
IGLOGIN WB2OSZ-5 123456
```

Contact the author if you can't figure out how to generate a passcode for your ham radio call.

If you want to transmit information from the servers, you need to specify the radio channel and the via path for the packet header. Example:

```
IGTXVIA 0 WIDE1-1,WIDE2-1
```

You might want to apply a filter for what packets will be obtained from the server. Example:

```
IGFILTER m/50
```

Read more about filters available here:

http://www.aprs-is.net/javaprsfilter.aspx

Finally, we don't want to flood the radio channel. The IGate function will limit the number of packets transmitted during 1 minute and 5 minute intervals. If a limit would be exceeded, the packet is dropped and warning is displayed in red.

```
IGTXLIMIT 6 10
```

7 Receive Decoding Performance and CPU Requirements

As Dire Wolf evolved over the years, a lot of effort has gone into continual improvement of the decoder performance. How do we measure this in an objective repeatable fashion? Some contrived lab test measuring bit error rates vs. signal to noise ratios doesn't replicate the real world problems that we encounter. The de facto standard is the TNC Test CD from WA8LMF. It contains recordings of almost 26 minutes with about 1000 packets.

Rather than running it real-time, the CD was "ripped" to files so we can process it quicker and easier. In the test below, we are using the same decoder that is in Dire Wolf, it's just repackaged into a separate application, called "atest," which reads from an audio file instead of the soundcard.

As the demodulator performance went up slowly, the amount of CPU power required rose quicker. To get from 963 packets decoded up to 988, the required CPU time more than doubled. The times below are for a Raspberry Pi, model B.

Decoder	atest options	Packets	Seconds	Comments
		decoded	to process	
		from Track 2		
Α	-P A	963	436	Original
В	-P B	964	572	
С	-P C	969	729	
D	-P D	921	226	This was fine tuned for 300 baud and not
				intended for 1200 baud.
E	-PE	988	894	New in version 1.2
F	-P F	963	275	Only for 1200 baud, 1200/2200 Hz,
				44100 sample rate.
				Was previous default for Raspberry Pi.

This is not an issue for a typical laptop or desktop computer which are about an order of magnitude faster and have floating point vector processing units that can perform multiple operations simultaneously.

For slow computers, a special case demodulator, called "F" (for fast), was added. Rather than being general to handle all cases, it was fine tuned to handle only one specific case. This was the default for the Raspberry Pi in earlier versions.

In version 1.2 we tried something new to compensate for the imbalance between the two audio tones. The separate document, "A Better APRS Packet Demodulator," explains this in detail. We could squeeze out a few more error-free frames but the CPU usage continues to climb.

Decoder	atest options	Packets	Seconds	Comments
		decoded	to process	
		from Track 2		
E+	-P E+	1008	981	
F+	-P F+	972	363	

CPU usage is getting rather high. For the "E+" case, Dire Wolf is taking about 2/3 of the CPU capacity just to receive. Additional bursts of processing are required to transmit and communicate with attached applications. The operating system needs to talk to the soundcard and other devices. It can't keep up.

We add a new option, "-D," to divide the audio sample rate by the specified integer. The amount of CPU power required drops dramatically without too much impact on the demodulator performance.

Decoder	atest options	Packets	Seconds	Comments
		decoded	to process	
		from Track 2		
E	-PE	988	894	Default rate.
E	-P E -D 2	985	348	Divide audio sample rate by 2.
E	-P E -D 3	977	237	Divide audio sample rate by 3.
E+	-P E+	1008	981	Default rate.
E+	-P E+ -D 2	1005	401	Divide audio sample rate by 2.
E+	-P E+ -D 3	1006	275	Divide audio sample rate by 3.
				** New default for version 1.2. **
E+	-P E+ -D 4	1001	231	Divide audio sample rate by 4.

It's a win-win situation. More packets decoded in less time.

You can change it, individually per channel in the configuration file. On a Raspberry Pi, the audio sample rate is now divided by 3 by default. The following are equivalent in version 1.2.

MODEM 1200 E+
MODEM 1200 E+ /3

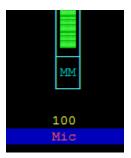
8 Troubleshooting

First check the audio gain on your input device. I've always found it necessary set it at the maximum or fairly high. In one case, with a new USB audio adapter, the initial microphone input gain was set to zero.

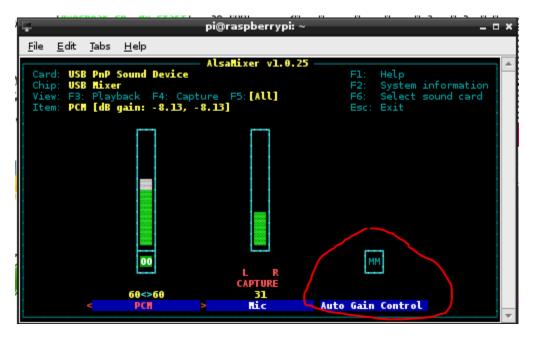
Different systems may have different applications for configuring the sound system. If using "alsamixer" follow this procedure:

- Press F6 to select the "sound card." Use up and down arrows and press Enter.
- Press F4 to set the Capture level. Use up arrow to set level to the max or fairly high.

Even if you crank up the gain to maximum, the input might be muted. Notice the "MM" here:



This means the input is muted. Press the "M" key to unmute. You might not be able to get the audio level input low enough if Auto Gain Control is enabled on the soundcard input. If you see something like this:



Be sure the Auto Gain Control shows "MM" which means disabled. If it shows "00," select it with the right/left arrow keys and press the "M" key so it displays "MM."

Did you run apt-get and rpi-update to get the latest software and firmware? After many months of aggravation, it finally worked right after an update mid October 2013.

8.1 Before that time...

The SignaLink USB worked OK but I couldn't get a cheap USB audio adapter to operate properly.

The same USB audio adapter worked fine on Microsoft Windows. It worked fine with Ubuntu Linux on a regular PC. It was a miserable failure on the Raspberry Pi.

Every 100 seconds, there was a debugging message similar to these in green:

```
ADEVICEO: Sample rate approx. 39.2 k, 0 errors, audio level CH0 ... ADEVICEO: Sample rate approx. 39.8 k, 0 errors, audio level CH0 ... ADEVICEO: Sample rate approx. 39.5 k, 0 errors, audio level CH0 ...
```

It reveals that the audio input system was providing less than 4000000 samples during a period of 100 seconds. We expect this to be close to 4410000 for the 44.1 kHz sample rate. Instead we find it is about 10% lower than expected. With the sample rate that far off, or 10% of the audio samples getting lost somewhere, nothing gets decoded. If no signals are decoded, look for these messages for a clue.

The USB audio device was fine on Windows and Ubunu Linux on a regular PC, so it was probably a software driver issue. After the most recent software/firmware update, it's now fine. The messages every 100 seconds show the expected number of audio samples for that time period.

If you can't copy any signals, make sure you have software/firmware no earlier than mid October 2013, crank up the audio input level, and then look for the debug messages every 100 seconds.

8.2 USB Hubs

Avoid putting a USB hub between the Raspberry Pi and the USB Audio adapter. I've received reports that exactly the same configuration works fine with the direct connection and a significant number of audio samples are lost when going through a hub. With direct connection this was observed:

```
ADEVICEO: Sample rate approx. 44.1 k, 0 errors, audio level CH0 ...
```

With the hub in the middle, numbers like this were seen:

```
ADEVICEO: Sample rate approx. 42.7 k, 0 errors, audio level CH0 ... ADEVICEO: Sample rate approx. 42.8 k, 0 errors, audio level CH0 ...
```

Some complain about these annoying messages but they provide valuable debugging information.

8.3 Use with AX25 kissattach

Sometimes kissattach has an issue with the Dire Wolf pseudo terminal. This shows up most often on Raspbian but sometimes occurs with other versions of Linux.

kissattach: Error setting line discipline: TIOCSETD: Device or resource busy Are you sure you have enabled MKISS support in the kernel or, if you made it a module, that the module is loaded?

Consult the *Linux AX25* section of the *User Guide* for more details and work-arounds.