Sigmira 1r13

Operations Manual

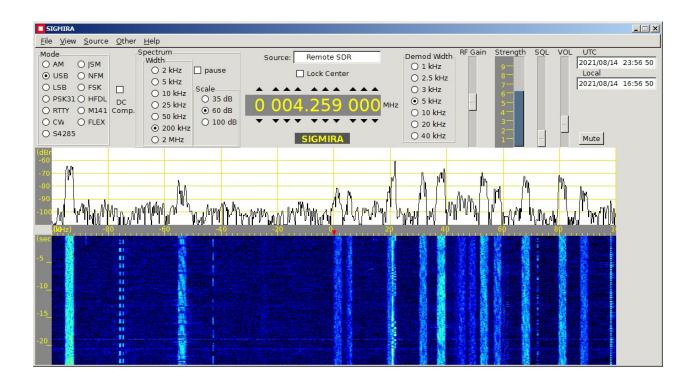


Table of Contents

1 INTRODUCTION	4
2 PROGRAM INSTALLATION	4
2.1 Windows Version	
2.1 WINDOWS VERSION	
3 CONTROLS AND DISPLAYS	5
3.1 WATERFALL DISPLAY	
3.2 Spectrum Display	
3.3 Rx Text Display	
3.3.1 File->Save/Save As	
3.3.2 Edit Menu	
3.4 Phase Plane Display	
3.5 Modes	
3.5.1 AM	
3.5.2 USB	
3.5.3 LSB	
3.5.4 PSK31	
3.5.5 RTTY	
3.5.6 FSK	
3.5.7 CW	
3.5.8 STANAG 4285	
3.5.9 HFDL	
3.5.10 JSM (Japanese Navy "Slot Machine")	
3.5.11 NFM	
3.5.12 FLEX Pagers	
3.6 DC COMP	
3.7 Spectrum Controls	
3.7.1 Spectrum Width	
3.7.2 Spectrum Base	
3.7.3 Spectrum Scale	
3.7.4 Spectrum Pause	
3.8 FREQUENCY.	
3.9 DEMODULATION WIDTH.	
3.10 Strength Indicator	
3.11 Squelch Control	
3.12 Mute	
3.13 FILE MENU.	
3.13.1 Open	
3.14 Source Menu	
3.14.1 SDR-IQ [™] , SDR-14, RTL-SDR	
3.14.2 SDR-IP	
3.14.3 WaveIn (Sound Card)	
3.14.4 Wave File	
3.14.5 SDRplay RSP Devices	
3.15 VIEW MENU	
3.16 CALIBRATE MENU	
3.16.1 Calibrate Frequency	
4 SIGNAL DATABASE	

4.1 SORTING	
4.2 Editing	
4.3 Fields	
4.1 SORTING. 4.2 EDITING. 4.3 FIELDS. 4.4 BUTTONS.	
4.5 Signal Database File Menu	54
5 SDR-IQ™ OPERATION	54
6 CONVENTIONAL RECEIVER CONNECTION	
7 OPERATION WITH NETWORKED RECEIVERS	
7 OPERATION WITH NETWORKED RECEIVERS 7.1 General	
7.1 General 7.2 Networked Receive Characteristics	
7.1 General 7.2 Networked Receive Characteristics	
7.1 General	
7.1 General 7.2 Networked Receive Characteristics 7.3 Editing the Network Receiver Database	

1 Introduction

Sigmira[™] is a Software Defined Radio (SDR) application program that runs on Windows. Sigmira offers wide band spectrograph and waterfall displays. It also demodulates various "ham" radio, commercial, and military signals.

It operates with the following signal sources:.

- RFSpace SDR-IQ and SDR-14
- RemoteSDR
- SDRplay
- Sound Card (WaveIn)
- Wav file
- RTL-SDR
- Any other via Virtual Audio Cables or VB cables

Its features include:

- "Waterfall" and spectrograph displays of up to 2 MHz width.
- Demodulation of HFDL, PSK31, FSK, SITOR-B, CW, NFM and Flex pager signals.
- Demodulation of NATO STANAG 4285.
- Demodulation of the "Japanese Navy Slot Machine" to QPSK symbols.
- Point and click signal selection/tuning.
- Signal database with automated tuning and logging.
- Playback of saved .wav files.
- Phase plane display.
- 16 bit sampling. Typical sampling rate: 48 ksps.
- Signal strength meter.
- Adjustable squelch.
- Clock display with UTC and local time.

The distribution package includes examples of HFDL, PSK31, RTTY, SITOR-B, CW, NFM, JSM, Flex pager, and STANAG 4285 signals in .wav files.

http://www.saharlow.com/technology/sigmira

Sigmira is Copyright 2021 by Steven A. Harlow.

2 **Program Installation**

2.1 Windows Version

If you already have your signal source hardware installed and working then the necessary drivers are already in place.

Sigmira is distributed in the executable self-installer 'sigtools_1r17_installer.exe'. To install Sigmira simply run the installer.

2.2 Receiver

The nicer features of Sigmira are available when it is used with the SDR-IQ, SDR-14, RTL-SDR, SDRplay, etc.. But it is not necessary to own receiver hardware to make use of Sigmira.

If you own an SDR device connect it to your PC via USB cable. If Windows asks for a driver for the device tell it to find the drivers in the <sigtools install root>/drivers directory. (The drivers are from Future Technology Devices International, Ltd. [FTDI] and are also available from their web site. <u>http://www.ftdichip.com/Drivers/VCP.htm</u>)

Instructions on connecting a conventional radio receiver are given in section 6. Connecting a receiver is not necessary in order to use Sigmira with .wav files. See section 8, Networked Receivers.

3 Controls and Displays

3.1 Waterfall Display

The waterfall display displays a time history of the signal spectrum. It is also used to select which signal to demodulate.

The waterfall display is the bottom third of the window. Higher frequency components are displayed to the right. The strengths of the frequency components are represented by brightness or color. The display shifts down with time so that the most recent signal content is displayed upper most. The display has a frequency scale along the top and a time scale along the left edge.

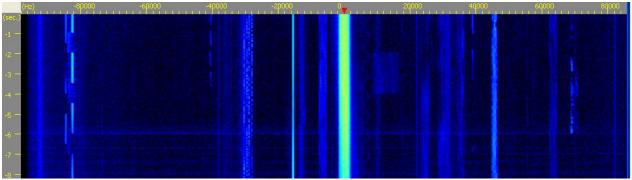


Figure 1 - Waterfall Display

The frequency scale is in Hz and is relative to the center tuning frequency. The frequency span of the waterfall display is adjustable with the "Spectrum" "Width" controls. There are also controls to pause the waterfall and to adjust the scaling of the display intensity level.

The scroll rate is adjustable. Move the cursor into the time scale on the left and roll the mouse wheel.

The waterfall display is also used to select which signal to demodulate. CW, PSK31, S4285, etc. are all narrow band transmissions. They show as bright vertical bands in the waterfall display. PSK31 and CW transmissions show as single vertical bands. An FSK transmission shows as a pair of vertical bands. Multiple transmissions may be present on the input and in the waterfall display at one time.

Select the particular transmission to be demodulated by right-clicking on it in the waterfall display.

A little red triangular pointer in the frequency scale indicates the frequency selected for demodulation of the signal. When the 'Lock Center' option is off the select frequency is shifted to the center of the display. When the 'Lock Center' option is on the display center frequency does not change and the red triangle moves to the selected frequency.

When in PSK31 or CW mode right-click directly on the signal. When in FSK mode two triangles are shown on the frequency scale. The triangle to the right is red and corresponds with the 'Mark' frequency. Right-click on the higher frequency column of the desired signal, i.e. the mark frequency. (It is easier to click on a signal when the displayed spectrum width is narrower.)

With the sound card or wave file selected as the source the maximum waterfall width is 10 kHz.

3.2 Spectrum Display

The spectrum display displays a plot of signal level vs. frequency.

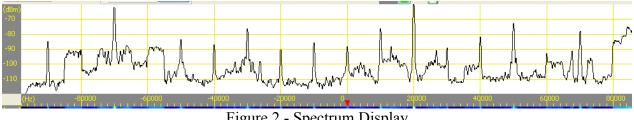


Figure 2 - Spectrum Display

There is a signal power scale at the left of the display. When SDR-IQ is selected as the source the scale is in dBm at the antenna input. When the source is the sound card or a wave file the scale is in dB relative to the full scale internal sample representation.

Tuning/frequency selection can be accomplished by right-clicking on the spectrum display just as on the waterfall display.

The base level of the display can be adjusted by rolling the mouse wheel while the cursor is in the spectrum display.

3.3 Rx Text Display

The received text display pops up whenever one of the CW, PSK31, FSK, RTTY, or S4285 mode buttons is clicked.

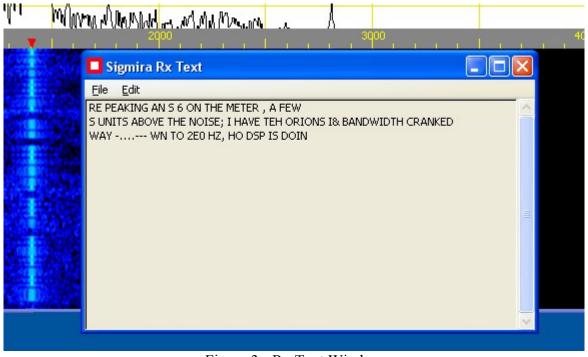


Figure 3 - Rx Text Window

The demodulated text of the selected transmission appears here. (RTTY, PSK31, CW, STANAG 4285, FSK, HFDL, and FLEX pager transmissions all originate as text type messages.)

Received text can be saved by following the File=>Save or File=>Save As menu dialogs. The text display can be cleared by selecting the Edit=>Clear Text menu item, Text can be selected with the mouse cursor and copied to the clipboard with Ctrl-C. (You may have to temporarily switch to a non-text demodulation mode [e.g. AM] so that the text stops moving while you do the selection.)

Whenever the demodulation mode is changed or the frequency is changed by 100 Hz or more a note is displayed in the Rx Text window.

Sigmira™ 1r13

3.3.1 File->Save/Save As

Selecting Save or Save As brings up a dialog window that can be used to save the received demodulated text to a windows text file. The entire text from the Rx Text window is saved to the file. The default name of the destination text file at start-up is 'untitled'. The name can be changed with the 'Save As' selection. Sigmira remembers any such change to the destination file name. Subsequent 'Save' selections will write to the remembered destination file name. Any save to a file overwrites the previous contents if any.

3.3.2 Edit Menu

The Sigmira Rx Text window Edit menu has one entry, 'Clear text'. Selecting 'Clear text' clears the received text display.

3.4 Phase Plane Display

The Phase Plane display can be brought up by selecting View->Phase Plane from the main window menu bar.

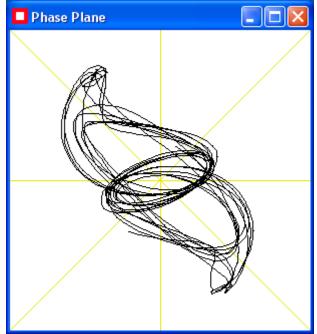


Figure 4 - Phase Plane Window

The phase plane displays the time varying phase and amplitude of the signal. The phase is relative to the tuning VFO. At any instant in time the phase and amplitude of the signal are represented by a point in the display. The amplitude is represented by the distance of the point from the center of the display. The phase is represented by the angle of the point around the center. Positive angle is counter-clockwise. To the right is 0 degrees, up is 90 degrees, left is 180 degrees, and down is 270 or -90 degrees. If a constant amplitude pure carrier of the same

frequency and phase of the VFO were being received it would be displayed as a dot to the right of the center. If the carrier were 1 Hz higher in frequency than the VFO the dot would revolve around the center counter-clockwise making one revolution per second.

A Binary Phase-Shift Keying (BPSK) High Frequency Data Link (HFDL) signal is shown in the figure above. In the absence of noise and distortion a BPSK signal would appear as two dots 180 degrees opposite each other with a line between them. Distortion in the radio wave propagation causes the wiggling seen above.

The phase plane display is appropriate for viewing phase modulated signals such as PSK31, STANAG 4285, HFDL, etc. Certainly it is also of interest out of general curiosity. And it is also useful as a tuning and calibration indicator.

An AM signal will generally appear as a line segment radiating from the center of the display. If Sigmira is tuned exactly to the carrier frequency the line will be stationary and not revolve around the center. As mentioned above, if the signal is 1 Hz higher in frequency the line will revolve around the center counter-clockwise at 1 Hz. This makes the phase plane a sensitive tuning indicator. This is useful in calibrating the SDR-IQ. See section 3.16.1.

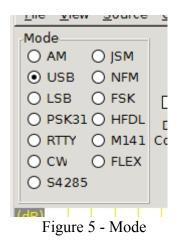
The phase plane display is functional in all of Sigmira's demodulation modes.

In the AM, USB, and LSB modes the Demod Width setting is effective. In those modes the signal content on both sides of the tuning frequency are processed into the phase plane display (i.e. USB and LSB modes do not cut off the sidebands in the phase plane display.) It is appropriate and advantageous to set the Demod Width to a minimum which includes the signal of interest and excludes outside noise. The Demod Width is the one-sided width. So, if one is viewing a phase modulated signal with a two-sided (total) width of 2 kHz, the Demod Width should be set to 1 kHz.

In the PSK31 and S42285 modes the Demod Width setting is not important. In addition, the demodulator locks phase with the signal and that is reflected in the phase plane display. With a good PSK31 signal the displayed phase will bounce between approximately 0 degrees and 180 degrees.

3.5 Modes

The Mode refers to the type of signal, AM, USB, LSB, CW, STANAG 4285, FLEX, etc. The mode setting selects which type of signal the program demodulates. The currently active mode is selected by clicking on the particular Mode 'radio button' in the 'Mode' group in the upper left corner of the program window.



3.5.1 AM

The AM is the usual Amplitude Modulation that everyone is familiar with. AM is typically used to send audio signals. The "AM broadcast band" (also know as MW or "medium wave") ranges from 520 to 1610 (1710 in the US) kHz. International short-wave broadcasters operate on several bands in the 3 to 30 MHz range. AM can also be described as double sideband (DSB) with carrier. AM signals can have various widths, 10 kHz and 20 kHz being common.

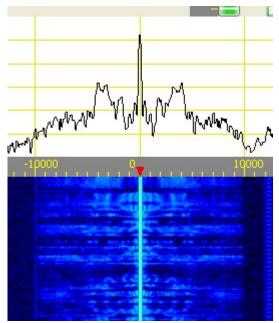


Figure 6 - Typical AM Signal

To tune in an AM transmission click on the AM radio button in the Mode group in the upper left corner of the Sigmira window. Then select the particular transmission to demodulate by right-clicking on it in the waterfall display.

For the optimum signal to noise ratio the demodulation width should be adjust to match the signal. See section 3.8 "Demodulation Width". For the signal in the above figure a demodulation width of 10 kHz would be appropriate.

AM broadcast signals have modulation bandwidths of 5 or 10 kHz, typically.

The squelch control can be used to reduce response to noise. See section 3.11, Squelch Control.

3.5.2 USB

USB stands for Upper Sideband. It is one of the two single sideband (SSB) modes. SSB is normally used to send audio signals. Normal amplitude modulation produces two sidebands, one on each side of the carrier. The audio information is in the sidebands and the sidebands are mirror images of each other. SSB was developed to be more efficient than AM. Only one sideband is transmitted, thus SSB signals are narrower. There are two advantages to that. 1. More signals can be packed into the same bandwidth and 2. the reduced receiver bandwidth allows less noise in with the signal. Typically the carrier is suppressed too which is an additional power savings at the transmitter. The down side to SSB is that demodulating the signal is more complex. The receiver must either reinsert a locally generated approximation of the carrier or use a complex signal and the Hilbert transform. And tuning is critical. If the receiver is tuned to high or low in frequency the resulting audio is shifted up or down giving what is described as a Donald Duck sound.

Since an AM signal has two sidebands there are two choices as to which sideband to transmit for SSB. Thus Upper Sideband (USB) and Lower Sideband (LSB). Hams make extensive use of SSB. By convention they use LSB below 10 MHz and USB above 10 MHz.

The USB or LSB receive modes can sometimes be used for improved reception of a regular AM signal. If there is an interfering signal that affects one sideband of an AM signal more than the other one can use the appropriate SSB mode to demodulate the sideband suffering the least interference. Another case where SSB receive mode can be used to advantage on an AM signal is when the AM signal is experiencing severe frequency selective fading. In instances where the carrier specifically is faded out the sound can be particularly distorted using the normal AM demodulation mode. Switching to SSB significantly reduces the distortion because SSB demodulation does not use the carrier.

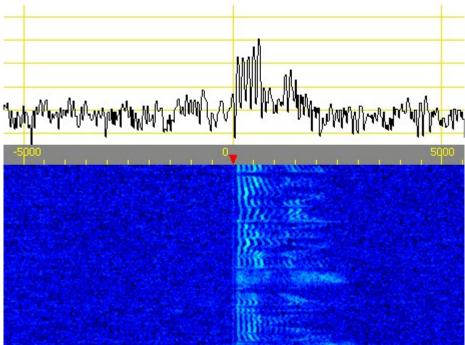


Figure 7 - Upper Sideband (USB)

3.5.3 LSB

LSB stands for Lower Sideband. It is the flip side of Upper Sideband, USB. See USB above.

3.5.4 PSK31

PSK31 is a narrow-band character-based modulation mode used by hams for interactive long distance communication in the HF bands. PSK31 employs Binary Phase Shift Keying (BPSK), Huffman coding, and a low baud rate. Those features are particularly suited to the HF bands which offer over-the-horizon and world wide propagation with the challenges of ionospheric fading and limited frequency resources.

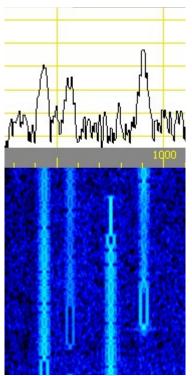


Figure 8 - PSK31 Signals

The popularity of PSK31 rather reduced now in 2020. PSK31 signals were regularly found at 7070 kHz and 14070 kHz. The lower frequency, 7070 kHz, is more active evenings and the higher frequency is more active middays. A PSK31 signal appears as a bright column in Sigmira's waterfall display. Figure 8, above, is an example of the appearance of PSK31 signals. That example is from a particularly active period but it is common that there are many signals visible at one time. The narrowband nature of PSK31 signals allows many signals to be placed close together, even within the passband of a short-wave communications receiver, and thus be visible at the same time at one receiver frequency setting.

To demodulate a PSK31 transmission click on the PSK31 radio button in the Mode group in the upper left corner of the Sigmira window. Then select the particular transmission to demodulate by right-clicking on it in the waterfall display. The demodulated text will appear in the Rx Text window.

Sigmira uses a narrow filter to separate the selected transmission from the rest of the input. The signal strength indicator indicates the strength of the filtered signal. The waterfall display continues to display the entire input.

The SDR-IQ has something of a "0 Hz birdie". Samples from the SDR-IQ have a DC offset that shows as a peak or faint line at the center of the spectrum/waterfall. It can interfere with PSK31 demodulation (when 'Lock Center' is off). The offset can be minimized with the 'DC Comp.' feature described below. However in some cases it may still be advantageous to tune slightly off from the PSK31 signal (so that it is not at the center of the spectrum), turn 'Lock Center' on, and then tune directly on the PSK31 signal.

The squelch control can be used to reduce response to noise. See section 3.11, Squelch Control.

3.5.5 RTTY

RTTY (Radio Teletype) is a narrow-band character-based modulation mode used by hams for interactive long distance communication in the HF bands. (Note that the Sigmira RTTY mode is specifically oriented to the ham FSK signal. For the general FSK demodulation mode see the FSK mode section, below.) RTTY employs Frequency Shift Keying (FSK) modulation, "Baudot" coding, and a low baud rate. Those features are suited to the HF bands which offer over-the-horizon and world wide propagation with the challenges of ionospheric fading and limited frequency resources. RTTY is one of the earliest forms of digital modulation having originated probably before the 1940s. As such it doesn't include the more modern features of Huffman coding and phase shift keying that PSK31 does.

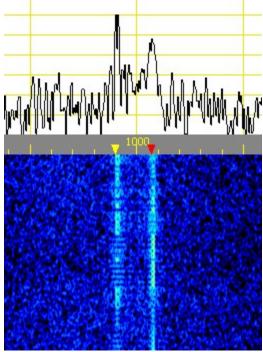


Figure 9 - RTTY Signal

RTTY signals can generally be found from 7080 kHz to 7100 kHz and 14080 kHz to 14100 kHz. The lower frequency range is more active evenings and the higher frequency range is more active middays. (When using an external receiver connected to the sound card input set the receiver to USB mode.)

The frequency shift for RTTY signals is 170 Hz. The higher frequency is the "mark" frequency and the lower frequency is the "space" frequency. The baud rate is 45.45. A RTTY signal appears as a pair of bright column in Sigmira's waterfall display. Figure 9, above, is an example of the appearance of a RTTY signal.

To demodulate a RTTY signal click on the RTTY radio button in the Mode group in the upper left corner of the Sigmira window. When in RTTY mode two triangles are shown on the frequency scale. The triangle to the right is red and corresponds with the 'Mark' frequency. Right-click on the higher frequency column of the desired signal, i.e. the mark frequency. (It is easier to click on a signal when the displayed spectrum width is narrower.) The demodulated text will appear in the Rx Text window. (When using a conventional receiver with Sigmira for RTTY demodulation make sure the receiver is in USB mode.)

Sigmira uses a narrow filters to separate the selected transmission from the rest of the input. The signal strength indicator indicates the strength of the filtered signal. The waterfall display continues to display the entire input.

The squelch control can be used to reduce response to noise. See section 3.11, Squelch Control.

3.5.6 FSK

FSK (Frequency Shift Keying) is a mature, uncomplicated "digital" mode. It is generally used for character-based communication. It originated before the 1940s and is still used by military, government, and private maritime entities for long-haul, over-the-horizon communication on the HF bands. Hams also use FSK. Sigmira has a separate RTTY mode that is specifically oriented to the ham FSK mode.

Sigmira's FSK mode can also be used to demodulate SITOR-B signals. SITOR-B is 170 Hz shift, 100 baud, CCIR 476 character set.

There are over a dozen entries for FSK signals in the Sigmira signal database. There are many FSK signals that run almost constantly. They are most commonly 850 shift and 50 or 75 baud. They are also typically encrypted. Some of the stronger signals received in southern California are 5345, 7593, 9085, and 14436.3 kHz (center). (We're in a sunspot low in 2020 so these HF signals are not so prominent.)

At least two of the SITOR-B signals heard in southern California are from NMC. They consist of maritime weather information (unencrypted). They are at 8416.5 and 12579 kHz (center).

To demodulate an FSK signal click on the FSK radio button in the Mode group in the upper left corner of the Sigmira window. When in FSK mode two triangles are shown on the frequency scale. The triangle to the right is red and corresponds with the 'Mark' frequency. Right-click on the higher frequency column of the desired signal, i.e. the mark frequency. (It is easier to click on a signal when the displayed spectrum width is narrower.) The demodulated text will appear in the Rx Text window. (When using a conventional receiver with Sigmira for FSK demodulation make sure the receiver is in USB mode.)

Sigmira's FSK mode offers user selectable shift, baud rate, mark/space inversion, and source format.

The FSK Control window is displayed whenever the FSK mode is selected (from the mode buttons in the upper left corner of the main window) or when called up via the View menu,

View=>FSK Control, also in the upper left corner of the main window. The FSK Control window appears as below:

FSK Control	
Shift (Hz)	
⊙ 170 ○ 250 ○ 420 ○ 650	○ 850 ○ 170
Baud Rate	
○ 45.45 ○ 50 ○ 75 ⊙ 100	O 100
Mark/Space Invert	Differential
Source Format	
O Binary O ITA2 O ASCII O CCIR4	76
Bits Per Character (ASCII only)	
● 8 ○ 7	
Parity (ASCII only)	
⊙ None ○ Odd ○ Even	
Stop Bits (ASCII only)	
⊙ 1 O 2	
LTRS FIGS	

Shift

The Shift radio buttons allow the demodulator to be set to match the frequency shift of the received signal. The shift of the received signal is most easily observed when the displayed spectrum width is set at lower values, i.e. 25 kHz or less. When in the FSK mode two triangles are shown on the frequency scale. They correspond to the 'mark' and 'space' frequencies. (Typically the higher frequency is the mark frequency.) Select the shift so that the separation of the triangles matches the separation of the displayed columns of the received signal. Then right-clicking on the mark column in the waterfall display tunes the demodulator so that the red triangle is over the mark frequency and the yellow triangle is over the space frequency. Most signals are either 170 or 850 shift. The FSK Control window offers preset buttons for several known shift values. In addition the far right shift button allows the selection of a shift value that can be entered by the user.

Baud

The Baud radio buttons allow the demodulator to be set to match the baud rate of the received signal. Most 850 shift signals are either 50 or 75 baud. Amateur RTTY signals are 45.45 baud (170 shift). SITOR-B signals are 100 baud (170 shift). Selection of the correct baud rate can be verified in the eye diagram described below. The FSK Control window offers preset buttons for the common baud rates. In addition the button on the far right allows the selection of a rate that can be entered by the user.

Mark/Space Invert

The Mark/Space Invert checkbox allows the swapping of which frequency is interpreted as mark and which as space. Normally the higher transmitted frequency is mark. A few stations transmit

mark as the lower frequency. In addition, when using a conventional receiver it is common for the receiver to switch to lower sideband mode below 10 MHz. That swaps the mark and space frequencies. The Mark/Space Invert checkbox can be used to adapt the demodulation to an inverted sender or to a lower sideband swap.

Differential

One way to avoid sensitivity to which of the two frequencies is the 'Mark' and which is the 'Space' is to send the signal 'differentially'. With a differential signal a change from one frequency to the other would be considered a mark (or a '1') and no change would be considered a space (or a '0'). This yields the original intended symbol values even if the two frequencies somehow become reversed somewhere in the signal path. The 'Differential' checkbox enables this interpretation. The 'Invert' checkbox still works in conjunction with the Differential feature. Invert changes whether a frequency change is considered a '1' vs a '0'. I don't know of any particular FSK signal that is differential but Sigmira is prepared for the possibility. (PSK31, which is not an FSK signal, is an example of a differential signal.)

Source Format

The Source Format radio buttons allow one to select the final interpretation of the data for display in the Rx Text window.

Binary

The 'Binary' mode is basically no interpretation. The individual bits are displayed. The binary mode is appropriate for viewing encrypted transmissions because the encryption conceals the original source format. The binary mode is also useful in assessing unknown signals and looking for patterns that indicate the actual format. The regular pattern of start and stop bits is usually apparent when viewing ITA2 data in binary mode.

ITA2 (Baudot)

The 'ITA2' mode prints data formatted at the source in the 5-bit start stop serial ITA2 alphabet. (ITA2 is sometimes referred to as "Baudot".)

ASCII

The 'ASCII' mode prints data formatted at the source in the common 7 or 8-bit start stop serial format. The control window also includes options for the number of bits, parity bits, and stop bits. Those settings are only effective in the ASCII mode. I've never seen any HF signal encoded in ASCII.

CCIR476

The 'CCIR476' format is used by SITOR-B and NAVTEX. (In CCIR476 characters are seven bit long with exactly four '1' bits per character. Each character is sent twice with the repeated characters being interleaved and delayed by five characters.)

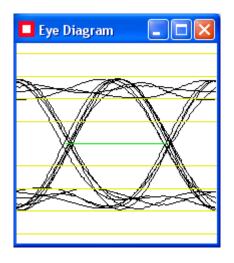
LTRS/FIGS

The LTRS and FIGS button allow the LTRS/FIGS state of the demodulator to be manually changed. Both ITA2 and CCIR476 use a LTRS/FIGS state to extend the range of characters that may be sent. The five bits of an ITA2 character and the seven bits of a CCIR476 character basically only allow for about 32 different characters. "LTRS" and "FIGS" control characters are

Sigmira™ 1r13

used to tell the demodulator to switch back and forth between two tables of meanings for the transmitted character. When in the LTRS state '10000' (ITA2) means 'T' and when in the FIGS state '10000' means '5'. The demodulator must track the LTRS/FIGS state. Noise, interference, and fading can cause the demodulator to miss or receive false LTRS and FIGS control characters and thus lose track. In that case a string of inappropriate letters or figures will be observed in the received text display. If the demodulator is erroneously in the FIGS state clicking the LTRS button forces the demodulator back to the LTRS state. Likewise the FIGS button forces the demodulator to the FIGS state. The LTRS and FIGS buttons in the FSK Control window are also effective when Sigmira is in RTTY mode.

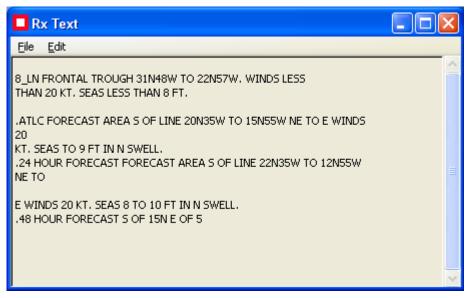
The eye diagram display is shown below.



The eye diagram allows the signal quality and the demodulated bit timing to be observed. With a good quality signal and the correct baud rate selected the diagram appears as above. There is one complete "eye" that is centered and "open". The signal zero-crossing points are at the ends of the green line. Note that ITA2 and ASCII signals can have irregular bit timing. The functioning of the eye diagram is adjusted to compensate for that when the ITA2 or ASCII source format is selected.

Two recorded examples of FSK signals are included in the Sigmira package. They can be found at <sigtools_install_root>/examples/fsk_8r432_090418_1236_ksm.wav and <sigtools_install_root>/examples/sitorb_8r432_090418_1504_ksm.wav. The FSK example is 1000 Hz center, 170 shift, 45.45 baud, ITA2. The SITOR-B example is 1000 Hz center, 170 shift, 100 baud, CCIR476. To demodulate the SITOR-B example open the .wav file using the "File" menu in the main Sigmira window. (Refer to section 8 of this document for more details on opening .wav files.) Set the tuning frequency to 1000 Hz in the frequency control of the main window. (Section 3.8 can be referred to for more details on frequency setting.) Select "FSK" in the "Mode" menu in Sigmira's main window. Then in the "FSK Control" window select the 170 Hz shift. Select the 100 baud rate. The Mark/Space Invert box should be unchecked. Set the

"Source Format" option to "CCIR476". The decoded text should appear in the Rx Text window as shown below.



The squelch control is usable in the FSK mode.

3.5.7 CW

CW is old fashioned Morse code over radio. "CW" stands for "continuous wave". It is a historical term arising from the fact that the very first transmitters of a hundred years ago could not always produce the long 'dash' sound.

CW is "low tech". It requires the least hardware. All one needs to transmit a CW signal is a radio frequency generator with an on-off switch. That simplicity plus a nostalgia factor account for the longevity CW. CW does place greater demands on the receiver software, which has traditionally been implemented in a human being.

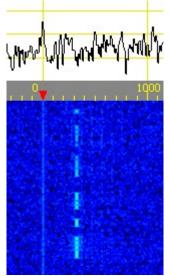


Figure 10 - A CW Signal

Interestingly, Morse code incorporates a basic concept that has been most recognized in the more modern Huffman coding. Huffman codes were invented some one hundred and twenty years later. In both codes the most common characters are mapped to the shortest transmit sequences.

CW signals can generally be found from 7000 kHz to 7070 kHz and 14000 kHz to 14070 kHz. The lower frequency range is more active evenings and the higher frequency range is more active middays.

A CW signal appears as a bright column in Sigmira's waterfall display. To demodulate a CW signal click on the CW radio button in the Mode group in the upper left corner of the Sigmira window. Then select the particular transmission to demodulate by right-clicking on the signal in the waterfall display. Initially Sigmira has to adapt to the rate at which the code is being sent. CW can be sent at a rate anywhere from 3 words per minute to 60 words per minute. While adapting Sigmira will typically print a number of 'E's in the Rx Text window. With a decent signal Sigmira will adapt to the correct speed within 10 seconds

Sigmira uses a very narrow filter to select the CW transmission out of the total input. The signal strength indicator indicates the strength of the filtered signal. The waterfall display continues to display the entire input.

(The CW filter is fixed at 50 Hz and is perhaps too narrow for some uses. If the automatic to-text deciphering is not need then CW can be listened to in the USB mode. A selectable range of CW filters will be added to a future version of Sigmira.)

Most of the SDR receivers have something of a "0 Hz birdie". Samples from the SDR have a DC offset that shows as a peak or faint line at the center of the spectrum/waterfall. It can interfere with CW demodulation (when 'Lock Center' is off). The offset can be minimized with the 'DC Comp.' feature described below. However in some cases it may still be advantageous to tune slightly off from the CW signal (so that it is not at the center of the spectrum), turn 'Lock Center' on, and then tune directly on the CW signal.

In CW mode the 'Squelch' control can optionally be used to set the signal level that is considered as "key down". When the signal level is above the squelch level it is considered a "key down" period, i.e. part of a 'dit' or a 'dah'. When the signal level is below the squelch level it is consider to be a "key up" interval, i.e. between a 'dit' or 'dah'. But it is not necessary that the best setting for this control be selected manually. If the squelch level is set at 0, at the bottom, the program automatically adjusts to what it calculates is the best level. Also see section 3.11, Squelch Control.

3.5.8 STANAG 4285

STANAG 4285 is a digital modulation mode designed by NATO for HF radio communications. It appears to be used mostly for long haul broadcast communication by various navies and maritime organizations. STANAG 4285 is a 2400 baud, 8-PSK signal.

The symbol rate is always 2400 symbols per second but various modulation options result in data rates from 75 to 3600 bps. The 2400 sps rate and PSK modulation result in a bandwidth of about 2400 Hz. Thus STANAG 4285 is wider than many HF data signals but still fits in the bandwidth of a typical voice signal. The signal has a 256 symbol frame structure which includes an 80 symbol synchronization pattern and three 16 symbol reference patterns. The sync and reference patterns along with 8-ary symbol scrambling facilitate adaptive channel equalization. Channel equalization is advantageous for a high data rate signal in the HF band considering the associated ionospheric fading. The sync pattern also facilitates Doppler adaptation as well as frame synchronization and symbol rate adaptation.

STANAG 4285 provides for forward error correction (FEC). FEC adds redundancy to the data such that the receiver can detect and generally correct errors in the received data. In STANAG 4285 FEC is implemented in a rate 1/2, constraint length 7 convolutional code and interleaver. The convolutional code effectively adds as many "parity" bits as there are data bits, thus the source data rate is 1/2 the coded data rate. Decoders for convolutional codes are less effective against bursts of errors. Bursts of errors are common in radio channels so an interleaver is typically employed along with a convolutional code. The interleaver shuffles the order of the data bits. A deinterleaver in the receiver reverses the interleaving. The errors of a bursts of errors caused by the radio channel are spaced out by the deinterleaver in the receiver.

The length of the interleaver in STANAG 4285 is a selectable option. Two nominal values are available, "short" and "long". The longer interleave is more resistant to channel noise bursts and fades. The length of the interleaver also corresponds with a real time delay in the data becoming available at the output of the demodulator. The "short" interleave corresponds with a delay of about 1 second. The "long" interleave corresponds with a delay of about 1 second. The "long" interleave corresponds with a delay of about 1 second. The "long" interleave corresponds with a delay of about 10 seconds. So if the "long" interleave option is selected any disturbance or change in the received signal won't be evident until 10 seconds later.

The STANAG 4285 specification includes an optional "message mode". The message mode is something of a digital squelch. The message mode specifies a "start of message" bit pattern (SOM) and an "end of message" bit pattern (EOM). When the message mode is selected the receiver is only to output data after a SOM is detected and before the next EOM is detected. In this observer's opinion the message mode is something of a misfeature. There are other less error prone methods of achieving the same result and no known actual users of STANAG 4285 use the message mode.

Sigmira employs an adaptive decision feedback equalizer (DFE) for channel equalization and a Viterbi decoder for the convolutional code.

STANAG 4285 transmitters have typically used a modem that produces an audio signal which is then the input to an USB transmitter. The carrier or center of the audio band STANAG 4285 signal is specified as 1800 Hz. The commonly quoted frequency for a STANAG 4285 radio signal is the tuning of the USB transmitter. So the actual center of the STANAG 4285 RF signal is 1800 Hz higher than the quoted frequency. Sigmira tuning is oriented to the center of the signal. When using Sigmira with the SDR-IQ the frequency setting should be the commonly

quoted frequency plus 1800 Hz. When using Sigmira with a conventional receiver tune the receiver to the commonly quoted frequency, USB mode, and set Sigmira to 1800 Hz.

Monitoring from southern California STANAG 4285 signals are often found around 6500 kHz and 8500 kHz. They are more common and stronger at night. A couple common signals are at 6400.5, 8490, and 8554.4 kHz (center). All observed signals have data rates of either 300, 600, or 1200 bits per second. A slight majority operate at 600 bits per second. All of them operate with the long interleave option.

Most of the observed signals appear to be encrypted. (The bit autocorrelation is flat and the run length statistics are the same as a Bernoulli process.) But there are signals transmitted in the clear. The French navy has been broadcasting several test signals in the clear for some time now. Examples are 8480.3, 8626.8, and 8647.8 kHz (center). The 8626.8 signal is at 600 bps. The others are 300 bps. All use the long interleave. They use the ITA2 alphabet.

The spectrogram of a STANAG 4285 signal is given in Figure 11, below. This is not a particularly strong signal. Significant fading is also evident. Even so, good copy was obtained from this signal.

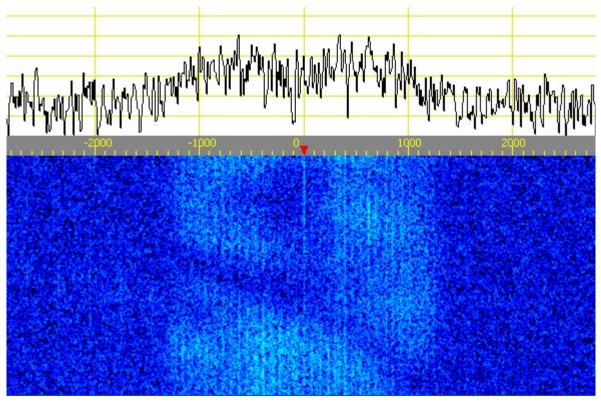


Figure 11 - STANAG 4285 Signal Spectrogram

To demodulate a STANAG 4285 transmission click on the S4285 radio button in the Mode group in the upper left corner of the Sigmira window. The Rx Text window and STANAG 4285 Control window should appear.

S4285 Control
Sync-
Data Rate (bps)
○ 75 ○ 150 ⊙ 300 ○ 600 ○ 1200 ○ 2400
Interleave
💿 Long 🔘 Short 🔵 None
Source Format
💿 Binary 🔘 ITA2 🔘 ASCII
Bits Per Character (ASCII only)
Parity (ASCII only)
💿 None 🔘 Odd 🔘 Even
Stop Bits (ASCII only)
⊙ 1 ○ 2
Message Mode
Sync Detect
💿 Optimal 🔵 Quicker

Figure 12 - STANAG 4285 Control Window

In the upper part of the control window are two indicators, the Sync indicator and the FEC Quality indicator. The Sync indicator illuminates green when Sigmira locks on to the synchronization of a STANAG 4285 signal. (Since the signal energy from 80 out of every 256 symbols contributes to sync detection it is possible detect sync even from signals which are too poor to yield good payload data.) The FEC Quality indicator is a thermometer type graph that indicates the quality of the demodulated data before the Viterbi decoder. If the indicator is above the 70% level the Viterbi decoder is able to correct virtually all errors. Good copy can still be obtained in the 50 to 60% range. (Note that the FEC Quality indicator is after the deinterleaver. If the interleave setting is changed to 'Long' or if the tuning is changed there will be a 10 second delay before the FEC Quality indicator gives a true indication.)

The Data Rate radio buttons in the control window allow the selection of the signal's expected data rate. It appears that virtually all real signals are either 300, 600, or 1200 bits per second. For an unknown signal one can determine the data rate by switching between the rates and observing the FEC Quality indicator. The FEC Quality indicator will be near zero for the incorrect rate. (Setting the data rate to a value above 600 bps also affects the deinterleaver. After changing the data rate to a value above 600 or back again one must wait the interleaver delay, 10 seconds for the 'Long' interleave, before the FEC Quality indication catches up.)

The Interleave radio buttons in the control window allow the selection of the signal's expected interleave. It appears that virtually all real signals use the long interleave. The interleave results in a real time delay through the demodulator. The short interleave results in a delay of about 1 second. The long interleave results in about 10 seconds. Changing the interleave setting resets

the deinterleaver and the interleave delay must pass before good data works its way through the deinterleaver again. The interleaver is also reset if the data rate is changed to or from any value above 600.

The Source Format radio buttons allow one to select the final interpretation of the data for display in the Rx Text window. The 'Binary' mode is basically no interpretation. The individual bits are displayed. The binary mode is appropriate for encrypted transmissions because the encryption conceals the original source format. Most signals appear to be encrypted. The binary mode is also useful in assessing unknown signals and looking for patterns that indicate the actual format. The regular pattern of start and stop bits is quite apparent when viewing ITA2 data in binary mode. The 'ITA2' mode prints data formatted at the source in the 5-bit start stop serial ITA2 alphabet. (ITA2 is sometimes referred to as "Baudot".) Several real broadcast test signals have been observed using ITA2. The 'ASCII' mode prints data formatted at the source in the common 7 or 8-bit start stop serial format. The control window also includes options for the number of bits, parity bits, and stop bits. Those settings are only effective in the ASCII mode. (But I've never seen a STANAG 4285 signal that used ASCII.)

The Message Mode check box controls the STANAG 4285 specified optional "message mode". When in the message mode the printing of demodulated data begins when the specified "start of message" bit pattern is detected and ends when the specified "end of message" bit pattern is detected. When not in message mode all data are printed. Message mode filtering is never applied when the 'Binary' Source Format is selected.

The Sync Detect radio buttons allow one to choose between two signal synchronization detector settings. With the 'Optimal' setting the sync detector is optimized for stable synchronization and minimal errors for both good and marginal signals per the STANAG 4285 specification. The 'Quicker' setting modifies the sync detector to synchronize instantly when given a good quality signal. Instant synchronization is not part of the mission of STANAG 4285. Generally the Sync Detect option should remain set to Optimal.

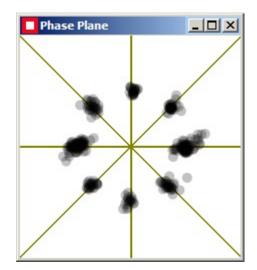


Figure 13 - Phase Plane Display for STANAG 4285 Signal

Figure 13 above illustrates the phase plane display when a high quality STANAG 4285 signal is being received. The dots correspond to the received symbols. The 8-PSK nature of the signal is quite clear. Note that at the lower data rates, including 300 and 600 bps, good copy is possible even with signals so noisy and distorted that an 8-PSK constellation is hardly recognizable in the phase plane. (The constellation in the phase plane is distinctly eight-pointed only when the signal is greater than the noise floor by 15 dB or more.)

A recorded example of a STANAG 4285 signal is included in the Sigmira package. To demonstrate Sigmira's STANAG 4285 operation begin by opening the example STANAG 4285 .wav file. The example is in the "examples" directory and is named "s4285_081020_2200_8r6268_600lita2_fnav_fum.wav". Wave files are opened using the "File" menu in the main Sigmira window. (Refer to section 8 of this document for more details on opening .wav files.) Set the tuning frequency to 1800 Hz in the frequency control of the main window. (Section 3.8 can be referred to for more details on frequency setting.) Select "S4285" in the "Mode" menu in Sigmira's main window. Then in the "S4285 Control" window select the "600" bps data rate. The "Interleave" option should also be set to "Long" (which is the default). Set the default). The signal will then be demodulated and wend its way through the deinterleaver and Viterbi decoder. After about 10 seconds text as shown below in Figure 14 will appear in the Rx Text window.

Rx Text	
<u>File E</u> dit	
	<u>^</u>
FAAA FAAA FAAA FAAA FAAA	
DE FUM FUM FUM FUM FUM	
TESTING TESTING TESTING TESTING	
RYRYRYRYRYRYRYRYRYRYRYRYRYRYRYRYRYRYRY	
0123456789876543210	
VOYEZ VOUS LE BRICK GEANT QUE J EXAMINE PRES DU GRAND WHARF	
FUM 5/6 5/8 5/12 5/16 5/22 5/0	
	~

Figure 14 - Text of the French Navy FUM STANAG 4285 Test Signal

3.5.9 HFDL

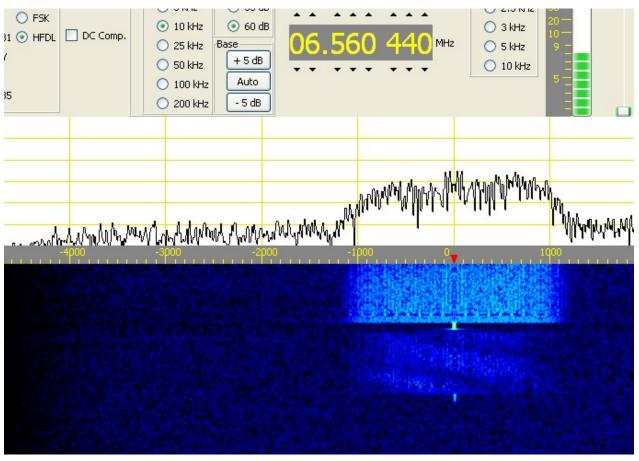
HFDL is a digital modulation mode used mainly by transoceanic aircraft. It was designed for and is employed on the HF bands because HF signals propagate over the horizon (unlike VHF, UHF,

etc.). HFDL messages often contain information such as aircraft ID, position (latitude and longitude), equipment status, and weather.

HFDL stands for High Frequency Data Link. It is specified in the document ARINC 635-3 by Aeronautical Radio, Inc. Generally HFDL is an 1800 baud BPSK signal. Occasionally it is also 4PSK or 8PSK.

The known frequencies on which HFDL is used are in the included Sigmira signal database. (See section 4 about the signal data base.) The frequencies can also be found in the internet. The commonly quoted frequencies are the "dial" frequencies. The center of the HFDL signal, i.e. the carrier, is 1440 Hz higher than the dial frequency. When using Sigmira with the SDR-IQ tune to the center frequency, i.e. the dial frequency plus 1440 Hz. (The entries in the signal database are the center frequencies.) When using Sigmira with a conventional receiver tune the receiver to the commonly quoted frequency, USB mode, and set Sigmira to 1440 Hz.

The spectrogram of a couple HFDL signal bursts is given in the figure, below.



HFDL Signals

To demodulate an HFDL transmission click on the HFDL radio button in the Mode group in the upper left corner of the Sigmira window. The Rx Text window will appear.

HDFL transmissions must be interpreted per the System Table. Transmissions contain indexes which are used to look up ground station names and frequencies in the table. At start up Sigmira obtains the system table by reading the file <sigtools_install_root>/bin/hfdl_system_table_51.txt. Version 51 is the current version as of November, 2020. Sigmira always reads the file with the highest version number. Updates are rare. HFDL demodulation still works without the table.

A recorded example of an HFDL signal is included in the Sigmira package. To demonstrate Sigmira's HFDL operation begin by opening the example HFDL .wav file. The example is in the "examples" directory and is named "hfdl_example.wav". Wave files are opened using the "File" menu in the main Sigmira window. (Refer to section 8 of this document for more details on opening .wav files.) Set the tuning frequency to 1440 Hz in the frequency control of the main window. (Section 3.8 can be referred to for more details on frequency setting.) Select "HFDL" in the "Mode" menu in Sigmira's main window. The signal will then be demodulated and the decoded text in the Rx Text window as shown below.

Rx Text	×
<u>File E</u> dit	
	^
MPDU	
Downlink	
NLP	
Ground Station ID 1 not synchronized SAN FRANCISCO - CALIFORNIA	
resreq	
medium priority slots 0	
low priority slots 3	
Unnumbered data:	
Performance Data	
perf data hfnpdu Flight Number ID UP0085	
Lat	
Long	
UTC04:05:48	
PD Version	
Vendor Specific	
Flight Leg 211 Ground Station ID	
Frequency	
Freq Search Count (prev) 0	
Freq Search Count (crnt)	
HF Data Disabled Time (prev) 12360	
HF Data Disabled Time (crnt) 6174 Uplink MPDUs Received	
1800 1200 600 300	
0 0 7 20	
Uplink MPDU Errors	_
Squitters Received	
Squitter Errors	
Downlink MPDUs Transmitted	
	=
1800 1200 600 300	-
0 0 1 2 Downlink MPDUs Acknowledged on first try.	
1800 1200 600 300	
0 0 1 2	
Frequency Change Codeno change	
Vendor Specific0x2	
	$\mathbf{\mathbf{v}}$

3.5.10 JSM (Japanese Navy "Slot Machine")

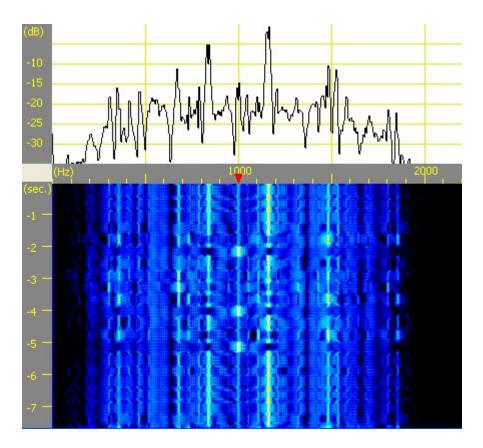
JSM refers to a rather strange signal. The consensus is that it is transmitted by the Japanese navy. When one listens to it in USB mode it normally sounds like a continuous repetitive melody.

Sigmira™ 1r13

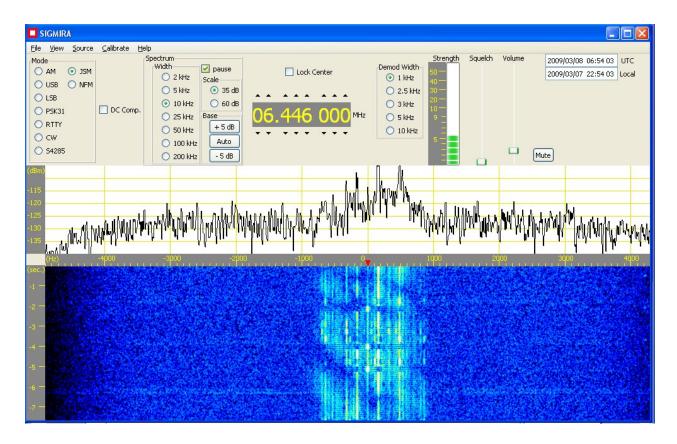
Some have likened the sound to that of a Las Vegas slot machine. Thus the name, JSM, Japanese Slot Machine.

In California it is commonly best received after 22:00 (local) and up to the hour after sunrise (local) on 6.446 and 8.589 MHz (center). It has also been heard on 4.154, 4.2325, 4.291, 6.251, 6.418, 8.314, 8.7045 MHz (center).

A recorded .wav file example of the signal is include in the Sigmira distribution package. It can be found at <sigtools_install_root>/examples/jsm_080904_0651_8r5888_example.wav. Sigmira displays the example as shown below:



A recent live capture is shown below.

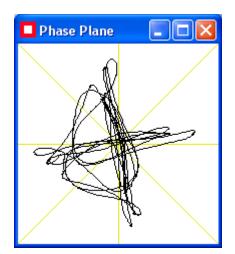


Most of the time the signal is an exactly repeating melody. It also includes an 11.4 Hz ticking sound. The melody repeats every 5.6 seconds. This is odd. Information theory tells us that the more predictable a signal is the less information it contains. This repeating signal contains basically no more than three pieces of information: a frequency, a phase, and a station ID. Occasionally the signal changes to become more chaotic and noise-like. That is the information payload. The melody intervals of the signal are called the "idle" time here.

The observed characteristics of the signal are consistent with a fixed (land based) broadcast to multiple, likely mobile, receivers over the horizon. The signals run continuously without interruption. The content of the signal on each of the various transmit frequencies appears identical. There is never anything like a "reply".

The actual frequency of the nominal 8.589 MHz signal has been measured at 8.588997 MHz +/-0.5 Hz. So the signal is not a top stratum frequency reference (such as NIST WWV).

A snapshot of the signal in the phase plane is shown in the image below. (The "Demod Width" was set to 1 kHz.)



Clearly it is a QPSK signal.

With some investigation the symbol rate was determined to be 1500.00 baud. The regular ticking sound was found to be an exactly repeating sequence of symbols. Clearly that serves as a channel probe and frame synchronization pattern. One tick sound period is here defined as a "frame". Frames were found to consist of 140 QPSK symbols. So the frame rate is 10.71428 Hz. The probe/sync pattern is 28 symbols which is one fifth the number of symbols per frame. It is found that, during the repetitive melody idle time, the remaining symbols of a frame consist of four repetitions of another 28 symbol pattern. So a frame appears to consist of five "blocks" of 28 symbols.

During idle time there is a finite set of symbol patterns that appear in the blocks. The patterns are designated here: ps, p0, p1, p2, p3, p4, p5, p6, p7, p8, p9, p10, and p11. The "ps" pattern is the probe/sync pattern. The rest are numbered roughly in the order of their frequency of occurrence. The "p0" pattern is most common.

The melodic idle time consists of an exactly repeating sequence of 64 frames. A 64 frame sequence is called a "super frame" here. The duration of a super frame is 5.973333 seconds. The frame/block pattern of the idle super frame is presented in the following table.

Frame Number	Block 0 Pattern	Block 1 Pattern	Block 2 Pattern	Block 3 Pattern	Block 4 Pattern
0	ps	p10	p10	p10	p10
1	ps	p10	p10	p10	p10
2	ps	p10	p10	p10	p10
3	ps	p0	p0	p0	p0
4	ps	p8	p8	p8	p8
5	ps	p8	p8	p8	p8
6	ps	p8	p8	p8	p8

Idle	Super	Frame	Structure
------	-------	-------	-----------

7	ps	p0	p0	p0	p0
8	ps	p9	p9	p9	p9
9	ps	p9	p9	p9	p9
10	ps	p9	p9	p9	p9
11	ps	p1	p1	p1	p1
12	ps	p11	p11	p11	p11
13	ps	p11	p11	p11	p11
14	ps	p11	p11	p11	p11
15	ps	p0	p0	p0	p0
16	ps	p6	p6	p6	p6
17	ps	p6	p6	p6	p6
18	ps	p6	p6	p6	p6
19	ps	p0	p0	p0	p0
20	ps	p5	p5	p5	p5
21	ps	p5	p5	p5	p5
22	ps	p5	p5	p5	p5
23	ps	p0	p0	p0	p0
24	ps	p4	p4	p4	p4
25	ps	p4	p4	p4	p4
26	ps	p4	p4	p4	p4
27	ps	p1	p1	p1	p1
28	ps	p3	p3	p3	p3
29	ps	p3	p3	p3	p3
30	ps	p3	p3	p3	p3
31	ps	p1	p1	p1	p1
32	ps	p11	p11	p11	p11
33	ps	p11	p11	p11	p11
34	ps	p11	p11	p11	p11
35	ps	p0	p0	p0	p0
36	ps	p7	p7	p7	p7
37	ps	p7	p7	p7	p7
38	ps	p7	p7	p7	p7

Sigmira™	1r13
Signina	1115

39	ps	p1	p1	p1	p1
40	ps	p3	p3	p3	p3
41	ps	p3	p3	p3	p3
42	ps	p3	p3	p3	p3
43	ps	p0	p0	p0	p0
44	ps	p0	p0	p0	p0
45	ps	p0	p0	p0	p0
46	ps	p0	p0	p0	p0
47	ps	p1	p1	pl	p1
48	ps	p0	p0	p0	p0
49	ps	p0	p0	p0	p0
50	ps	p0	p0	p0	p0
51	ps	p1	p1	p1	p1
52	ps	p0	p0	p0	p0
53	ps	p0	p0	p0	p0
54	ps	p0	p0	p0	p0
55	ps	p0	p0	p0	p0
56	ps	p1	p1	pl	p1
57	ps	p1	p1	pl	p1
58	ps	p1	p1	pl	p1
59	ps	p0	p0	p0	p0
60	ps	p0	p0	p0	p0
61	ps	p0	p0	p0	p0
62	ps	p0	p0	p0	p0
63	ps	p1	p1	pl	p1

The choice of which frame in the super frame should be designated as frame 0 was somewhat arbitrary. The numbering in the above table is used as the convention in this document.

The melody arises from the regular repeated simple patterns of symbols/phases. The p10 pattern is all one phase. So it produces a single tone. The p11 pattern is also all one phase but 180 degrees from p10. The other patterns are somewhat more complex and result in different and multiple apparent tones.

There are times that the signal departs from the idle pattern and sounds more "noisy". That is no doubt the data payload. There are typically from 4 to 20 such data intervals per hour. There seems to be no pattern to the minute of the hour at which data intervals begin. The length of each data interval is an exact multiple of super frames. At the end of a data interval the idle pattern resumes exactly in step and alignment as if there had been no interruption. So there appears to be an underlying continuous 64-frame super frame structure to the signal.

It appears that each data interval begins at frame 31 or frame 32 in the super frame (by the frame numbering convention here). Likewise the signal returns to idle at frame 31 or 32. There appear to be several formats for data intervals. They can roughly be characterized by the number of frames consisting of the fixed 'p' patterns (described above) that are mixed in with frames containing the data. The data interval super frames generally consist of a repeating pattern of four frames, some of which are random (data) symbols with the rest being the fixed patterns. The number of data frames in those four frame groups can vary from one to four. Data super frames in which less than all four frames of the four frame groups are data could be called "partial" data super frames. The full data super frames are often preceded and followed by partial data super frames.

The suspected data symbols have been extracted and analyzed via autocorrelation. The symbols were also converted to bits according to an expected Gray code. The autocorrelations have been flat indicating that the symbols are further coded and/or encrypted.

The abrupt boundaries of the data super frames could expose evidence of convolutional coding or interleaving if used. Such evidence has not been obvious in the observed signals.

Sigmira uses the 'ps' block of the signal frames to accomplish carrier recovery, symbol synchronization, frame synchronization, and channel equalization. The carrier reference phase was somewhat arbitrarily chosen such that the 'p10' pattern symbols are at 90 degrees. (0 degrees is to the right in the Sigmira phase plane display. 90 degrees is up.)

Sigmira displays the demodulation results in the "Rx Text" window. An example is given in the figure below.

File	<u>E</u> dit			
4	p11	p11	p11	pll
15	p0	pO	pO	pO
16	p6	p6	p6	p6
7	p6	p6	p6	p6
8	p6	p6	p6	p6
19	рO	рO	рO	pO
20	p5	p 5	p 5	p5
21	p5	p5	р5	p5
22	р5	p5	р5	p5
23	р0	рO	рO	рO
24	р4	p4	р4	p4
25	р4	p4	р4	p4
26	р4	p4	р4	p4
27	р1	p 1	р1	pi
28	р3	р3	р3	p3
29	р3	р3	р3	p3
30	р3	р3	р3	p3
31	р1	p1	р1	pi
32	р1	p 1	р1	+
3	.++000**+	+*00++*0+0*0)+.0 +++.*0+0+*	'oo*.*o*+**o.*o** 0+.*.+0+*o*0+.*+*o*.+*+++.*+ .*++00+0+00.*o*+0++0000++
4	***0++0.	00*+*+0+000.*.	.+ +*+.*00*.0*+	0+.00000**++++.*++0*.*+**0.*+00+.++0.*+*0++0*0.+*0000.++.0.00+
35	р0	рO	рO	рO
36	р1	p 1	р1	+*.+*+o**
37				0+0+0*.**+. 0**+*++0*0*+0++00*0+.*+. +++.+**.+*0+.0*+*.+*0.00
38	.0+0.+**+0	0+0+00+0+**+	++++ 0+.0.0.0*0	o*+o+**o**.*o.*+o* *+.**+.*o.+.+***o.+*+oo +.*+.o+*+*+oo.ooo*o*+
39	рO	рO	рO	рO
ŧO	р1	p 1	р1	+
1)*0.+.*+.* .*+0.+*.+*0*0**0*+0+0*+.* +*+0+*+0+0000*+0*.*+++.
ł2	.**00*+++	o.*+**+o.o+*-	+.0 0+0.0+0*+0	0+*0*++**.++0*+* *.+0+0*000**+*0.0+++*.+* 00+.*+0.*+.00.*.*000.+*
ł3	рO	рO	рO	pO
14	D 1	D1	D 1	···+··································

The above display resulted from demodulating the included example file: <sigtools_install_root>/examples/jsm_080904_0651_8r5888_example.wav. (Set the tuning frequency to the center of the signal, which is 1000 Hz for the example file.) The above display shows a transition from an idle interval to a "partial" data interval.

The display consists of one frame per line.

The first number in each line is the number of the frame in the super frame. Note that Sigmira must first detect the idle super frame pattern in order to synchronize the frame numbers. Before that occurs the frame numbers are displayed as "-1".

The rest of each displayed line/frame consists of either the recognized fixed group patterns or the raw QPSK symbols. Each received 28 symbol group is compared to the fixed patterns. If the group exactly matches a pattern, e.g. p0, p1, p2, etc., then the pattern number is displayed. Otherwise the 28 raw QPSK symbols are displayed. The four possible QPSK symbols are translated to printable characters for display. The translation is given in the following table.

QPSK Symbol Phase (degrees)	Printed Character
0	11
90	'+'
180	'o'
270	1*1

It is likely that two bits are encoded in each QPSK symbol using a Gray code. An arbitrary alignment could have been chosen and the resulting bits displayed in the Rx Text window. Any misalignment of the code would have been manifest in every bit or every other bit being inverted from the correct sense. Patterns in the data bits might be evident despite the inversions, but autocorrelations of the data so far have revealed no patterns. It was decided to simply display the QPSK symbols in order to reduce the volume of characters sent to the display.

3.5.11 NFM

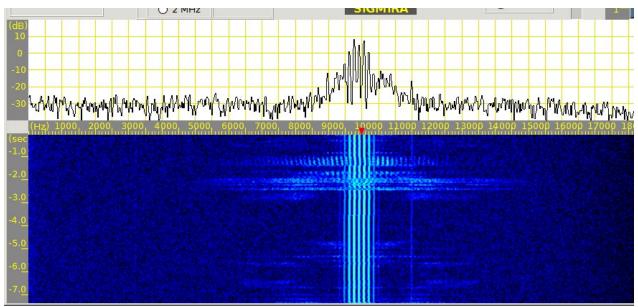
The NFM is narrowband frequency modulation. NFM is used for business, public service, ham, and GMRS voice communication. (Business and public service use is described as "Land Mobile Radio" [LMR].) It is primarily used on VHF and UHF.

NFM is similar to broadcast radio FM but much narrower in bandwidth and much lower in fidelity. NFM could be described as "toll-quality", i.e. following the original standards for land line telephones. The frequency response is basically 300 to 3000 Hz. The TIA-603-E standard is a good technical reference. NFM is very rarely used on HF.

Sigmira's NFM mode is optimized for NFM as just described. So normally it is best to select the 3 kHz "Demod Width" when using NFM. (A wider 5 kHz width might give somewhat better sound quality with the NOAA weather broadcasts. The NOAA signals are a little wider than the standard 3 kHz. They can be found around 162.4 to 162.55 MHz.)

NFM can occasionally be heard in the 10m ham band around 29.6 MHz. On rare occasions studio links originating in Texas can be heard in southern California around 25.99 MHz and 26.19 MHz. NFM is used in the 2 meter ham band, 144 to 148 MHz. NFM public service communications can be heard around 856 MHz, 860 MHz, and various other frequencies.

An example of the appearance of an NFM signal in Sigmira's spectrum and waterfall displays is given in the figure immediately below.



NFM Signal

The above display is from the included recorded example. It can be found at

<sigtools_install_root>/examples/nfm_210817_0839_2m_keller_net_crop_08.wav When playing the wave file set the demodulation mode to NFM, set the frequency to 10 kHz, and the demodulation width to 3 kHz. The recording is of a conversation on a 2m ham repeater. Looking at the spectrum, the central bars are from a Continuous Tone-Coded Squelch System (CTCSS). The voice information is in the sidebands that extend further out from the center.

The Sigmira NFM implements the de-emphasis described in TIA-603-E and not the "75 μ s" or "50 μ s" associated with the common broadcast FM.

Set Sigmira's tuning frequency to the center of the signal. It is not critical that the tuning be at the exact center.

The squelch control is also functional in NFM mode. (See section 3.11, Squelch Control.)

3.5.12 FLEX Pagers

Pagers are palm sized radio receivers with which one can receive short text messages. The message might be the phone number of someone who would like you to call them back. Pagers peaked in popularity in the years just before cellphones became practical. There is still a solid niche for pagers. The short message function is still quite useful for communication with hospital staff and emergency responders. Pagers are also lower cost than cellphones.

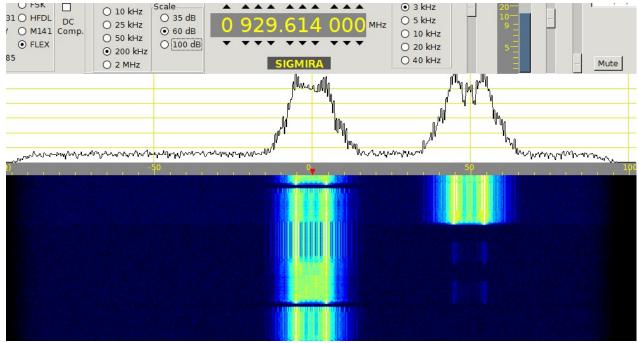
POCSAG was the first technical standard for pagers. It is still dominant in Europe from what I understand. You can find POCSAG signals in the US but in my area 95% of the traffic is FLEX.

Most of the traffic I see is in the 929.0125 to 929.9875 MHz range with additional signals in the 931.0125 to 931.9875 MHz range.

FLEX is transmitted as 2FSK and 4FSK. The shift frequencies are +/- 4800 Hz and +/- 1600 Hz. The symbol rates are 1600 baud and 3200 baud. Overall the signal has a width of about 24 kHz. Message information is formatted in blocks and frames in cycling time slots. A block is made up of 32 bit "code words". The code words are made up of 21 data bits, ten BCH error correction bits, and one even parity bit.

There are various types of formatted messages such as "Numeric", "Hex/Binary", "Alphanumeric", "Secure", and "Short". Each message is addressed to a specific pager. The "capcode" identifies the intended receiver. The capcodes are displayed with the messages in Sigmira's Rx Text window.

Some types of messages can be split across multiple frames. Messages are mostly not encrypted. A minority seem to be encrypted or in a special character set. Some transmitters seem to include extra optional features in the signal, such as a frequency reference preamble or a short frame ending tone.



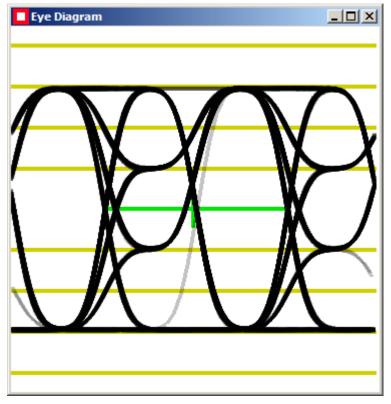
The spectrogram of a FLEX pager signal is shown below.

FLEX Pager Signal Spectrogram

To demodulate FLEX transmissions click on the FLEX radio button in the Mode group in the upper left corner of the Sigmira window. The Rx Text window will automatically appear. Tune to the approximate center of the signal. When receiving a live I/Q signal the tuning is not critical.

You can be as much as 12 kHz off center and Sigmira will adapt. It may take a couple frames to adapt. The "Demod Width" setting has no effect in the FLEX mode.

The squelch level can be adjusted so that the demodulator is active only when actual signal energy is present. This is beneficial but not strictly necessary for demodulation. The squelch setting can also give a cleaner eye diagram display.



Eye Diagram for FLEX Pager Signal

The figure above illustrates the eye diagram display when a good quality FLEX signal is being received. The signal shown is a 4FSK, 3200 baud FLEX signal. There are four symbol times shown, left to right. The continuous stream repeatedly writes over the symbol times resulting in multiple symbols being painted in each position. The four FSK frequencies are represented in the vertical levels. The symbol immediately to the left of center contains the "A phase" and the "B phase". The symbol immediately to the right contains the C and D phases.

A recorded example of a FLEX signal is included in the Sigmira package. To play and demodulate the example click on File=>Open in the Sigmira main window. Then navigate to the examples directory. (My apologies that the file menu is unnecessarily complicated. It looks like you have to click 'bin' and then above that 'sigmira_1r13'. Then the 'example' directory appears.) Select the file "flex_929614_210816_1854_10kshift_selected.wav" in the 'examples" directory. Set the frequency control to 10 kHz. The recorded example is offset by 10 kHz. Don't forget to

select the FLEX mode in the main window. As the demodulation proceeds you will see the FLEX messages displayed in the Rx Text window as in the image below.

Rx Text _ 🗆 🗙 <u>File Edit</u> >>> FIW cycle 13 frame 42 Standard Numeric capcode 8034250 41222 [22] Short Message capcode 7473908 35200000 Alphanumeric capcode 7850662 ASSGN RTM: 53 F Isolation:* Standard Dest Room-Bed: 455 Dest Bed Status: Stat Origin Unit: * M - ED Admitting: Level of Care: Car diac Telemetry Diagnosis: FLUID OVERLOAD CHEST PAIN R/O ACS [87] Alphanumeric capcode 7319261 53 F Isolation:* Standard Dest Room-Bed: 455 Dest ASSGN RTM: Bed Status: Stat Origin Unit: * M - ED Admitting: Level of Care: Car diac Telemetry Diagnosis: FLUID OVERLOAD CHEST PAIN R/O ACS [39] >>> FIW cycle 13 frame 43 Alphanumeric capcode 7897681 From: @pvhmc.org Subject: - STATUS CHANGE: BED: Tele 6 C684 Bed B BED ID: 06842 STATUS: Dirty STATUS TIME: 08/16/2021 18:52 [61] Alphanumeric capcode 7929710 From: @pvhmc.org Subject: - NEW JOB: BED: Tele 6 C684 Bed B BED ID: 06842 • CTATUS Dity [6/1]

FLEX Messages Demodulated from the Included Example Wav File

3.6 DC Comp.

The DC Comp. feature adapts to and cancels the DC offset in the samples from the various signal sources.

DC Comp.

Figure 15 - DC Comp. Button

Samples from most SDR hardware boxes have a DC offset. The DC offset shows as a peak or faint line at the center of the spectrum/waterfall. It can interfere with the demodulation of CW and phase modulated signals. The offset varies with RF gain setting and tuning.

In normal operation Sigmira subtracts an offset from each sample. That offset is a saved value (per source) that is fixed when the 'DC Comp' feature is off. When the 'DC Comp' feature is turned on Sigmira monitors the input samples and adapts the saved offset value to the current sample stream.

The different signal sources have different characteristics in their DC offset.

The SDR-IQ is relatively well behaved. Once adapted and canceled it basically doesn't change unless you change frequency by more that say 100 kHz or change the RF gain. Normal usage would be to leave the DC Comp feature off so that is not disturbed by actual signals. If the DC offset becomes annoying tune to a quiet area of spectrum in the neighborhood of the signals of interest. Turn the DC Comp on. The adaptation takes 5 or 10 seconds. The results are observable in the spectrum/waterfall particularly at the narrow width settings. When the central ridge is minimized turn DC Comp off.

The RTL-SDR I have (with an R820T tuner) seems to have a small DC offset. It can often be ignored. It can also be handled as described above for the SDR-IQ.

The SDRplay RSP1 does some compensation for the DC on its own. You can see it adapt after RF gain changes or large frequency changes. But it is not complete. The Sigmira DC Comp control can still be used for better compensation. It is reasonable leave the DC Comp feature clicked on the whole time while using the SDRplay. Beyond the DC offset the SDRplay also seems to have uncancellable low frequency noise of about 80 Hz width.

3.7 Spectrum Controls

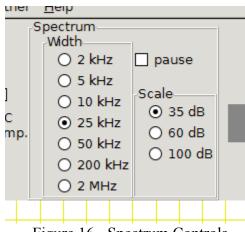


Figure 16 - Spectrum Controls

3.7.1 Spectrum Width

The Width buttons control the width of spectrum displayed in the spectrum and waterfall displays. The maximum supported width for the SDR-IQ/14 is 200 kHz. The maximum supported width

for the RTL-SDR and SDRplay RSP1 is 2 MHz. When the selected source is the sound card or a wave file the maximum width available is 25 kHz.

3.7.2 Spectrum Base

The spectrum base can also be adjusted by rolling the mouse wheel while the cursor is in the spectrum display.

3.7.3 Spectrum Scale

There are three buttons, "35 dB", "60 dB", and "100 dB" that control the scale of the spectrum and waterfall displays. Scale here refers to how bright a signal appears in the waterfall display or how high above the noise floor it is drawn in the spectrum display. Adjusting the display scale is necessary because absolute signal levels vary considerably.

At the smaller scale value smaller signals will be more prominently displayed. Note that the signal level scale in the spectrum display tracks the scale adjustments so that the indicated levels remain calibrated.

3.7.4 Spectrum Pause

The Pause button causes the spectrum, waterfall, phase plane, and eye diagram displays to pause. Demodulation, audio output, and all other functions continue.

3.8 Frequency

The Frequency indicator displays and controls the signal tuning frequency. The radix point is at the MHz position. The least significant digit is at single Hz. There are digits to display up to 9,999.999999 MHz but each receiver/signal source has its own maximum frequency limit.

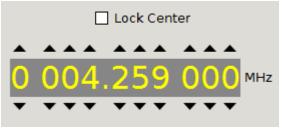


Figure 17 - Frequency Control

The displayed frequency is the frequency pointed to by the red triangle in the waterfall frequency scale. Positioning the PC mouse cursor over a digit and spinning the mouse wheel causes the frequency to roll up or down in that digit position. Clicking on a digit in the frequency display

causes the tuning frequency to be rounded in that digit place. (i.e. clicking on the 5 in a display of 10.500000 causes the frequency to change to 11.000000.) Clicking on the triangle above a digit causes the frequency to increment in that digit. Similarly, clicking on the triangle below a digit causes a decrement.

Clicking on a digit in the frequency display causes the tuning frequency to be rounded in that digit place. For example, clicking on the '3' in a displayed frequency of "121.378" will round the frequency to "121.000".

As described in the section on the waterfall display, right-clicking in the waterfall display (or the frequency scale, or the spectrum display) causes the tuning frequency to change to the frequency of the point clicked. If the Lock Center box is not checked then the center of the displayed spectrum range changes to that frequency. Consequently the red triangle will always be at the center of the display.

If the Lock Center box is checked then the center frequency of the displayed spectrum is fixed. When the tuning frequency is changed the red triangle will move to the corresponding location in the frequency scale. This is useful when the center frequency has been set to be within a group of signals of interest. The tuning can be changed from signal to signal and the red triangle will follow while the signals maintain their positions in the spectrum display.

3.9 Demodulation Width

The demodulation width control adjusts the bandwidth used in demodulating the signal. Signals vary in width. It is advantageous to adjust the demodulation width to be wide enough to include the signal but no wider so that no extra noise is included.

Demod Width	
○ 1 kHz	
○ 2.5 kHz	
⊖ 3 kHz	
⊖ 5 kHz	
⊙ 10 kHz	
○ 20 kHz	
⊖ 40 kHz	

Figure 18 - Demod Width

The demodulation width control is effective in the AM, USB, LSB, and NFM modes. AM and SSB signals have considerable variability in width. In the other modes (particularly CW, PSK31, FSK, RTTY, S4285, HFDL, and FLEX) the signals have defined widths and the demodulators are tailored to those widths.

The selectable widths refer to the demodulated output. AM signals are double sideband. So the transmitted width of an AM radio signal is twice that of the modulation. The transmitted radio signal for AM with 5 kHz modulation will be 10 kHz wide as seen in the spectrum/waterfall display. For such a signal the 5 kHz demodulation width would be appropriate. For SSB signals the transmitted width is the same as the modulation width.

AM broadcast signals have modulation bandwidth of 5 or 10 kHz typically. Ham SSB signals are typically 2.5 to 3 kHz wide.

3.10 Strength Indicator

The Strength indicator is a thermometer-like display that indicates the strength of the demodulated signal.

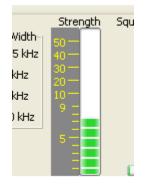


Figure 19 - S-Meter

If an SDR receiver is selected as the source then the strength scale is similar to a traditional S-Meter. The tick marks are "S-Units" up to "S9" and then dB above S9 above that. (Only S5 and S9 are marked with digits to save clutter.) S9 is -73 dBm or 50 μ Vrms at the antenna input.

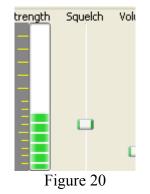
If the sound card or .wav file is selected as the source then the strength scale is relative. (It is not possible to accurately calibrate sound wave samples to radio signal strength. Recorded .wav files can have a wide range of recording levels. And an external receiver connected to the sound input typically has an AGC which results in a relatively constant audio level.) The strength scale is logarithmic. Full scale corresponds to the full strength representable by sound sample values.

Besides general curiosity, the Strength indicator is useful in adjusting the squelch (see section 3.11, Squelch Control) and setting up the optimum input level (see section 6, Conventional Receiver Connection).

3.11 Squelch Control

The squelch controls when the demodulation of the incoming signal is active. It sets a minimum strength level that the signal must reach in order to be demodulated. This is useful in preventing the background noise between transmissions from being interpreted as signal.

The squelch is a slider control that is controlled by clicking on it and sliding it up or down.



The squelch control is located next to the signal strength indicator and their levels correspond. If the indicated signal strength is higher than the squelch the demodulation will be active. If the squelch control is set to the bottom of its range the demodulation will always be active.

In CW mode the squelch control has a somewhat different function. In CW mode (Morse mode) the squelch control can optionally be used to set the signal level that is considered as "key down". When the signal level is above the squelch level it is considered a "key down" period, i.e. part of a 'dit' or a 'dah'. When the signal level is below the squelch level it is consider to be a "key up" interval, i.e. between a 'dit' or 'dah'. But it is not necessary that the best setting for this control be selected manually. If the squelch level is set at 0, at the bottom, the program automatically adjusts to what it calculates is the best level.

3.12 Mute

The Mute button controls whether the demodulated audio is played out to the PC sound output.

3.13 File Menu

The File menu is in the menu bar at the top of Sigmira's window. The File menu has two entries, Open and Quit.

🗖 SIGI	MIRA 1	rO	
<u>File</u>	ource	<u>C</u> alibrate	<u>H</u> elp
Ope	·		Spe
Quit (Ctrl+Q		
F :	01	E1. M.	

Figure 21 - File Menu

3.13.1 Open

Selecting Open brings up a dialog window that can be used to open an audio file. The file would typically be previously recorded undemodulated signal as from the short-wave receiver output. Once opened Sigmira will play and demodulate the signal just as if it were live input.

The file can be a .wav format file with any normal sample rate from 8000 to 48000 Samples/sec.

Several example .wav files are included with Sigmira. They are useful in demonstrating Sigmira's operation. See section 8, 'Wave File Input'.

3.14 Source Menu

Sigmira can operate with input from several different signal sources. The source is selected with the 'Source' menu at the top left of the main window.

MIRA				
<u>V</u> iew	<u>S</u> ource)		
₽ JSB (JSB (SB (PSK31 ((TTY (W (4285	<u>W</u> ave Reser Remo R <u>T</u> L-S	14 rved In und Card File rved oteSDR	i)	rum 2 kHz 5 kHz 10 kH 25 kH 25 kH 200 k 200 k
++	igure 22		Mer	

3.14.1 SDR-IQ[™], SDR-14, RTL-SDR

If the device is connected operation will start immediately when selection is made.

3.14.2 SDR-IP

The SDR-IP is not supported at this time.

3.14.3 WaveIn (Sound Card)

If "WaveIn" is selected the signal input is taken from the sound card or other Windows "Wavein" device. That may include "Virtual Audio Cables" if any are installed.

Often the sound card is used as the source when Sigmira is used with an external conventional receiver. The receiver's audio output is connected to the sound card input. See section 6 of this document.

Virtual Audio Cables, VB Cables, or similar utilities allow you to pipe the output from another SDR program into Sigmira. The other program must be set to output the audio to the virtual audio cable. Sigmira can then be used to further demodulate signals that the other program may not demodulate. Examples would include modes such as STANAG 4285, FSK, HFDL, etc.

As soon as you select WaveIn as the source the WaveIn Select window pops up. That allows you to select the specific input among the available inputs. It also allows you to select the sample rate and whether the input is I/Q. The WaveIn Selection window is also accessible via the Other=>WaveIn Select menu.

🗌 WaveIn	Selection			
Selected	Waveln Devi	ce		
Sound	Mapper			
Sound M	1apper			
Line 1 (\	/irtual Audio (Cable)		
CABLE	output (VB-Au	udio Virtual		
				 ▶
		Flip I/Q	Sample Rate	
			⊙ 44.1 ksps ○	192 ksps
			C	Cancel Accept

WaveIn Selection Window

The currently selected input is shown at the top. You can select a different input to use by clicking on it in the table.

I/Q

Many SDR programs offer output in I/Q (in-phase and quadrature) format. This is advantageous in that it provides twice the bandwidth and angle modulated signals can be processed without a frequency shift and offset. If the source program provides I/Q format click the I/Q checkbox. Sigmira will show 0 Hz in the center of the spectrum display and "negative" relative frequencies will be shown to the left. In this configuration you will not have to tune the source program such that the signal appears entirely in the positive half of Sigmira's spectrum display. Even if the signal extends into the negative frequency range you can right click to set Sigmira's tuned frequency to the reference center of the signal. Demodulation can then proceed.

Flip I/Q

Normally the in-phase signal is expected from the left channel and the quadrature signal is expected from the right channel. Some SDR programs reverse that. When the source and Sigmira don't match the spectrum display will be reversed with the lower frequency signals toward the right. This checkbox allows you to switch and match to what the other SDR source provides.

Sample Rate

Except for broadcast FM and television signals all other signals can fit within the bandwidth provided by the 44.1 ksps sample rate. The 192 ksps sample rate allows for a wider range spectrum display. The VAC utilities appear to automatically adapt to the sample rate requested. The actual bandwidth of the data provided is limited to the internal rate of the utility. The free version of one VAC utility is limited to 48 ksps. So when you select 192 ksps it provides 192 ksps but the information in those samples is limited to the equivalent of 48 ksps.

Note that when a WaveIn source is selected Sigmira's frequency controls do not control the tuning of the external source.

3.14.4 Wave File

Playback begins when a .wav file is opened with the File->Open selection. (Opening a wave file with File->Open also automatically sets the Source selection to Wave File.) Also see section 8.

3.14.5 SDRplay RSP Devices

Sigmira offers limited direct support for the SDRplay RSP devices. Sigmira will control the first tuner and will only use the first antenna input. Sigmira will also take the output from SDRuno via virtual audio cables. So you can use SDRuno to have full control over any RSP device and use Sigmira for further signal demodulation. See section 3.14.3 in this document.

For direct support Sigmira works with the versions 3.06, 3.07, and hopefully future versions of the SDRplay API. Install SDRuno with the API in order to have the API installed and available.

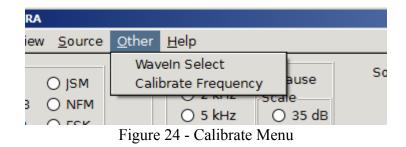
3.15 View Menu

Sigmira has several displayable sub-windows. They include the Signal Database, Phase Plane, S4285 Control, and Rx Text displays. They are described in sections 3.3 and 3.4. The View menu can be used to cause these sub-windows to be viewable.

SIG	MIRA				
<u>F</u> ile	<u>V</u> iew	<u>S</u> ource	<u>O</u> ther	Н	lelp
Mod	Sig	jnal <u>D</u> ata	base		ectrum-
OA	Ph	ase Plane	e		Mdth
οι	54	285 Cont	rol		○ 2 k⊦
_		K Control			○ 5 kH
OL	_				O 10 k
OF	<u>R</u> x	Text			-
OF	Ey	e Digram			💿 25 k
	Ne	twork Da	tabase		O 50 k
00				-	
0	1205				○ 200
	Figu	ıre 23 - V	'iew Me	nu	

3.16 Calibrate Menu

Under the top level "Other" menu there is a menu item that allows for calibration of the SDR-IQ crystal frequency.



3.16.1 Calibrate Frequency

The SDR receiver devices operate with a quartz crystal oscillator frequency standard. Crystal oscillators are highly accurate but have some small frequency offset. That offset is evident with current communications technology. Fortunately it is easy to compensate for the offset.

🗖 SDR-IQ Xtal Calibra	ation 🛛
5080000.0	Indicated Tuning
6080000.0	Actual Tuning
Factory Defau	lt]
	<u>o</u> ĸ

Figure 25 - Frequency Calibration

The SDR-IQ Xtal Calibration window allows one to adjust Sigmira to cancel out the frequency offset (for any of the SDR receiver devices). When the window comes up both the 'Indicated Tuning' and 'Actual Tuning' are set to the current indicated tuning. If either one is known to be incorrect the correct value can be entered. When the 'OK' button is clicked Sigmira will adjust so that thereafter the indicated frequency will match the actual.

As an example of how this can be done: One would tune to a known frequency standard such as WWV. The higher the frequency the more accurate the calibration. Bring up the phase plane display. (See section 3.4.) The signal should appear as a dot or line segment radiating from the center of the display. If Sigmira is tuned exactly to the carrier frequency the dot/line will be stationary and not revolve around the center. If the signal is higher in frequency the dot/line will revolve around the center clockwise. Adjust the tuning frequency so that the dot/line motion is the minimum possible. At this point the actual frequency will be most correct. Bring up the calibration window. Correct the displayed 'Actual Tuning' to be the known correct frequency of the reference signal and click 'OK'. (Note that the best practically achievable accuracy is about 0.5 Hz and that the crystal reference can easily drift several Hz in an hour.)

The 'Factory Default' button allows the calibration to be reset to the original nominal value.

4 Signal Database

The Signal Database allows one to log, categorize, recall, and instantly tune to signals. The number of entries is unlimited. A database already containing over a thousand entries is included in the Sigmira package.

The Signal Database is brought up by clicking the menu View=>Signal Database in the Sigmira main window. The Signal Database will appear as shown in Figure 26, below.

Frequency N 5.768500 a 5.772642 n 5.784000 fr 5.786800 fr 5.800000 a	mfsk fsk fdpsk am	Description voice scrambling 090210_1625 utc 0825 lcl rare short mfsk bursts 090115_2036 utc 1236 lcl fsk, s3 070210_0755 12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	can Cat1 fixed fixed unkn fixed	Scan Cat.	Freq Lookup Parameters
Frequency N 5.768500 a 5.772642 n 5.784000 fr 5.786800 fr 5.800000 a	Mode andvt mfsk fsk fdpsk am	Description voice scrambling 090210_1625 utc 0825 lcl rare short mfsk bursts 090115_2036 utc 1236 lcl fsk, s3 070210_0755 12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	Cat1 fixed fixed unkn fixed	Scan Cat.	
5.768500 a 5.772642 n 5.784000 fr 5.786800 fr 5.800000 a	andvt mfsk fsk fdpsk am	voice scrambling 090210_1625 utc 0825 lcl rare short mfsk bursts 090115_2036 utc 1236 lcl fsk, s3 070210_0755 12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	fixed fixed unkn fixed	Scan Cat.	Parameters
5.772642 n 5.784000 fr 5.786800 fr 5.800000 a	mfsk fsk fdpsk am	rare short mfsk bursts 090115_2036 utc 1236 lcl fsk, s3 070210_0755 12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	fixed unkn fixed		
5.784000 fr 5.786800 fr 5.800000 a	fsk fdpsk am	_ fsk, s3 070210_0755 12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	unkn fixed		
5.786800 fr 5.800000 a	fdpsk am	12 carriers, 200 Hz spacing, 101218_1552 utc 0752 lcl, Russia n CIS-12/AT-3004D? numbers, A3 (AM morse code), probably Cuba, nightly 0630UT	fixed		
5.800000 a	am	n CIS-12/AT-3004D?			
	am				
		C, s9+10 090104_1016 lcl	numbers		
5.801500 f		Havana? odd, 500 shift 40.5 bd, 091127_0629 utc 2229 lcl, 2 50s 75b 091215_2216 lcl	curious		500,40.5
5.805600 u	usb	Harmonics out ot 30 kHz on both sid es. Square wave? No., also show qsb, Havana again?, s9 0901 20_0549 utc 2149 lcl	curious		
5.807000 js	jsm	"slot machine" reported	maritime	jsm	
5.809000 u	usb	carrier, s9+15 qsb 070210_0800	unkn	unkn	
5.810000 u		8fd psk double sideband 090830_0555 utc 2255 lcl, again, Hav ana? 100701_2229 lcl	curious		
5.821000 n	mfsk	rare 090502_0251 utc 1951 lcl, 500s 40b 140928_0725 lcl	curious		
5.835000 a	am	music, s6 070210_0800	broadcast	voice	
5.845000 d	drm	drm 101218 1554 utc 0754 lcl	broadcast		
5.850000 a	am	broadcast strong religious 050919_2332	broadcast		
5.853000 u		? ami, multiple, male, "recovering" "banner" "manual code" "22 " "33" "track 377" "kilo"090514_0550 utc 2250 lcl	curious		
5.860000 a	am	broadcast, middle eastern?, s5 081226_0810	broadcast		
5.878600 li	link11	091224_0120 utc 1720 lcl	unkn	link11	

Figure 26 - Signal Database

Each signal entry in the database is one row. Right-clicking on a row tunes Sigmira to the frequency shown in Frequency field for the selected entry. The right-click also sets the demodulation mode according to the Mode field.

4.1 Sorting

Each entry has six fields: Frequency, Mode, Description, Cat1, Cat2, and Parameters. The six fields make up the six columns of the view. Left-clicking on the heading of a column causes the database view to be sorted based on that field. Clicking a column heading the first time causes an ascending sort. Clicking a second time causes a descending sort. If a signal entry is already selected when a sort is performed then that signal entry will be scrolled to the center of the view.

4.2 Editing

All of the fields of an entry are editable. Left-click on a field twice in order to edit it. Note that edits made are not saved to disk until Save or Save As is clicked in the Signal Database window File menu.

4.3 Fields

The Frequency field is in MHz. Whenever a new database entry is created Sigmira automatically enters the currently tuned frequency.

The Mode field corresponds with the demodulation modes in the upper left of the Sigmira main window. The Mode field is not case sensitive. "am", "AM", "aM", and "Am" are all equivalent. If the content of the Mode field doesn't correspond with one of Sigmira's recognized modes Sigmira defaults to USB. Whenever a new database entry is created Sigmira automatically enters the currently selected mode.

The Description field is available for any descriptive information the user would like to associate with the signal. Whenever a new database entry is created Sigmira automatically enters the string "new" plus the current date and time. This may of course be edited.

The Cat1 and Cat2 fields offers a means to place signals in user defined sortable categories. Clicking on the head of the Cat1 or Cat2 column will sort the database based on the contents of that field. So if all of the "numbers" station signal entries have the string "numbers" in the Cat1 field then sorting on the Cat1 column will display the numbers stations grouped together.

The contents of the Cat1 field are also interpreted to give a background tint to each entry. Ten predefined values for the Cat1 field are recognized. The values and the associated colors are given in the table below. The categories correspond roughly with the categories in the US Frequency Allocation Chart, <u>http://www.ntia.doc.gov/osmhome/Chp04Chart.pdf</u>.

Cat1 field value	Tint
av	light blue
broadcast	yellow
cb	light yellow
curious	pink
fixed	brown
ham	green
maritime	blue
misc	gray
numbers	red
ref	light gray

The Parameters field provides a means to associate additional demodulation parameters with a signal. The Parameters field will be fully implemented in a future version of Sigmira. The parameters can include things such as bit rate and interleave for STANAG 4285 signals or shift and baud rate for FSK signals, etc.

4.4 Buttons

Clicking the Freq Lookup button in the upper right portion of the Signal Database window causes the entry in the database closest in frequency to the current main tuning to be scrolled to the center of the database window and highlighted.

Clicking the Delete button in the upper center portion of the Signal Database window causes the current selected entry to be deleted.

Clicking the New button in the upper left portion of the Signal Database window causes a new entry to be created. The Frequency field of the new entry is automatically filled in with the current tuning frequency as shown in the Sigmira's main window. The Mode field is automatically filled in with the current mode. The Description field is filled in with the string "new" plus the current date and time.

4.5 Signal Database File Menu

In the menu bar at the top of the Signal Database window is a File menu. The File menu has four entries, Open, Read, Save and Save As.

Selecting File=>Open brings up a dialog window that can be used to open a signal database file. At startup Sigmira automatically opens the signal database file "sigmira.sdb". Using the Open dialog a different file can be opened. The contents of the newly specified file become the operating signal database. The newly specified file name also becomes the default for the Save function.

Selecting File=>Read brings up a dialog window that can be used to read in an additional signal database content from a file. The contents of the file are added to the current operating signal database. The file name specified in the Read operation does not change the default file name for Open and Save operations.

The signal database can be saved to disk by following the File=>Save or File=>Save As menu dialogs in the Signal Database window. The Save operation saves the signal database to the default signal database file name. The default file name at start-up is "sigmira.sdb". With the Save As operation the database is saved to a newly specified file. The newly specified file name becomes the default file name. Subsequent 'Save' selections will write to the remembered destination file name. Any save to a file overwrites the previous contents if any.

On exit Sigmira backs up the current database contents to the file "sigmira_bu.sdb" regardless of other operations.

5 SDR-IQ[™] Operation

Connect the SDR-IQ[™] to the computer with the USB (Universal Serial Bus in this case) cable. Connect an antenna to the SDR-IQ[™]. Start Sigmira and select the SDR-IQ[™] as the signal source via the 'Source' menu at the top left of the window.

6 Conventional Receiver Connection

Sigmira can operate with a conventional short-wave receiver. The receiver must have and must be operated in the Single Sideband (SSB) mode. Upper Sideband (USB) mode is preferable.

Connect the audio output of a short-wave receiver to the audio "Line In" audio input of the PC.

If you are fortunate your receiver will have an output that is before the volume control. In that case the PC receives a good constant level signal regardless of the receiver's volume control setting. On some receivers such an output is labeled AFSK Out. Otherwise you can tap the speaker output, set the volume for a moderate level, and avoid making changes to the level.

Adjust the gain and volume controls so that the signal isn't so strong that it becomes clipped or distorted. The controls on the receiver and also the PC's Recording Control window can be use to make adjustments. The PC sound controls can be accessed by clicking start=>run... and typing "sndvol32" or they can be accessed through the control panel by clicking start=>Control Panel=>Sound and Audio Devices=>Audio=>Volume. The resulting window will likely look like:

🚺 Master Volume								
Options Help								
Master Volume	Wave	SW Synth	Aux	Video	CD Audio	Line In	Microphone	Phone Line
Balance:	Balance:	Balance:	Balance:	Balance:	Balance:	Balance:	Balance:	Balance:
	▶ ◀	♦ ◀	♦ 4	♦ 4	▶ _ 4	● 4	▶ _ 4	▶ _ 4
Volume:	Volume:	Volume:	Volume:	Volume:	Volume:	Volume:	Volume:	Volume:
Mute all	Mute Mute	Mute Mute	Mute Mute	Mute	Mute	Mute Mute	Mute Advanced	Mute Mute
SiS 7012 Wave								

Figure 27

(It is a good idea to mute the "Line In" in the Master Volume window.)

Then the "Line In" input needs to be selected in the Recording Control.

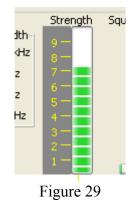
To get to the Recording Control window select Options=>Properties=>Recording. You should get a window like:

Recording Co	ntrol				
Options Help					
Stereo Mix	Aux	Video	CD Audio	Line In	Microphone
Balance:	Balance:	Balance:	Balance:	Balance:	Balance:
Volume:	Volume:	Volume:	Volume:	Volume:	Volume:
Select	Select	Select	Select	✓ Select	Select
SiS 7012 Wave					

Figure 28

Click the Select box under Line In and then adjust the Volume slider.

The optimum volume level setting is obtained when even the strongest input does not "peg" Sigmira's Strength indicator. Start Sigmira, select "Sound Card" source, and select the AM mode. Tune in a strong signal (of any type) on the receiver. Observe the Sigmira Strength indicator and adjust the volume so that the strength indicator never goes to full scale.



7 Operation with Networked Receivers

7.1 General

There are various SDR receivers that are accessible via the internet. There are various protocols or interfaces associated with these receivers. Sigmira currently supports the RFSpace/MoeTronix

RemoteSDR/SDRanywhere protocol. Currently there are about ten receivers that are accessible. They are located mostly in Europe and North America. A live list with their status is accessible at http://sdranywhere.com/cloud/sdrweb.py. More on RemoteSdrClient and SDRanywhere can be found here http://sdranywhere.com/cloud/sdrweb.py. More on RemoteSdrClient and SDRanywhere can be found here http://moetronix.com/remotesdrclient.html and here http://www.sdranywhere.com/SDRanywhere/SDRanywhere.html.

To use a networked receiver first bring up the Network Database window. Go to the main window and select the menu item View->Network Database. The "Network Connections" window is then displayed as illustrated below.

Network Co	onnections						1
<u>F</u> ile <u>E</u> dit							
Connection						Connect Time (minute	s)
Connecte	d to CI000160. Limit 25	54 minutes left.				1	
Connect	Disconnect						
Name	Location	Server Type	Net Address	Port	Status	Comments	Ī
CI000182	0.0, -180.0	CloudSDR/IQ	193.86.25.215	8023	private	Vetrnik AM	ſ
	0.0, -180.0	RemoteSDR	23.24.126.45	50001	private	Default Server	l
CI000017	63.21, 7.72	RemoteSDR	85.93.251.107	8024	private	OJS (private)	l
CI000025	0.0, -180.0	RemoteSDR	213.152.32.108	50000	private	I just can't seem to remember its settings and I have given up resetting them	
CI000141	45,11	RemoteSDR	88.149.150.61	50000		IK2YCW DX engeenering vertical antennas, beverages and 4 SQ - see webpage on your right	
CI000157	0.0, -180.0	RemoteSDR	194.24.239.54	35000	private	Default Server	I
CI000158	52.0, 13.0	RemoteSDR	87.187.154.215	50000	private	DM0HVL	İ
CI000160	47.00, -120.414	RemoteSDR	47.42.138.201			Ellensburg/Kittitas, WA CloudIQ (Ant1) Wellbrook Loop ALA15 30LNPro Imperium (Ant2) BigIR Vert 80m KC7JLU/AJ7JA Clo ud Mode	
CI000271	35.6971 N, 82.5607 W	RemoteSDR	71.85.130.68	50000		Wellbrook Loop North Carolina	
CI000290	48.49 ,9.74	RemoteSDR	62.68.25.29	50000		###### DL2NI, JN48UL, CloudIQ, active vertical antenna 10 kHz-30 MHz, k=-10dB, (ant1) and active loop 100kHz-30 MHz, (ant2) #######	
CI000297	0.0, -180.0	RemoteSDR	77.68.245.231	50000		Default Server	
CI000341	0.0, -180.0	RemoteSDR	89.202.124.3	50000		Default Server	
CI000343	0.0, -180.0	RemoteSDR	66.177.165.84	50000		1923 Beach	
HM000222	63.21, 7.72	RemoteSDR	85.93.251.107	8023	private	DXLC Smøla 290 deg ant #1	ĺ
IV001224	0.0, -180.0	RemoteSDR	24.158.242.249	50000		K4BPM Wellbrook AL-1530 Loop	Í
S001770	0.0, -180.0	RemoteSDR	185.89.55.189	50000		Karelia MW DX	
LG002339	51.949, 5.235	RemoteSDR	80.61.58.20	8019		*** PD0SWL/Netherlands/SDR-IQ/DXE-RPA1/Fan-Dipole ***	
MW003369	63.21, 7.72	RemoteSDR	85.93.251.107	8021	private	DXLC Smøla 290 deg ant #2	i
d							ĥ

The background tint of each entry has the following meaning:

- gray The entry is either the one currently selected or is from the local sigmira.ndb file.
- light yellow The entry is from the online database.
- darker yellow The entry is from the online database but the receiver/server is either private or currently occupied.

To connect to a networked receiver:

- 1. left-click on the entry
- 2. left-click on the "Connect" button.

Sigmira then goes through the connection process. The status and progress are displayed in the "Connection Status" text box. On a successful connection the status box will show something like:

Connected to: CI000160

Sigmira then receives, displays, and demodulates signals as it does from other source but with a few differences.

7.2 Networked Receive Characteristics

There is a detectable delay through the network connection and due to the compression used to reduce network bandwidth usage. There is about a second delay from changing the tuning frequency until the result is heard in the audio.

For Sigmira's data modes (PSK31, CW, S4285, FSK, etc.) Sigmira puts the networked receiver in USB mode. The typical steps to demodulate a data signal such as S4285 would be:

- Set the Spectrum Width to 10 kHz.
- Tune such that the data signal fits in the upper sideband passband of the receiver. That is, right-click at the left edge of the signal so that the signal falls in the 300 to 3000 Hz range of the upper sideband. The center of an S4285 signal would then be at about 1500 to 1800 Hz in the displayed spectrum.
- Lock the main tuning center frequency with the "Lock Center" check box in the main window.
- Then in the spectrum display right-click at the center of the S4285 signal. (A S4285 signal has a divot at the center of its spectrum that is visible if the fading is not too severe.) The center frequency of a S4285 signal is typically a number like xxx.000, xxx.200, xxx.400, xxx.500, or xxx.800 MHz.
- Select S4285 mode. The Sync indicator should turn green. It may be necessary to bump the frequency up or down for best results. Don't forget to uncheck the "Lock Center" box when done decoding a particular data signal and tuning to a different signal.

(Yes, all that is a bit of an inconvenience. I could make it automatic. Maybe in a future version.)

7.3 Editing the Network Receiver Database

Network receiver entries can be added and edited. At this time there isn't really a need to manually add entries The contents are read from the online database. You can click on Edit=>Refresh in the Network Connections window to update the displayed information. That will also update the displayed available/occupied status of the receivers.

One might add an entry via the Network Connections window Edit->New Connection menu item. Added entries will have the gray background.

Entries can be edited by selecting the entry (left-click) and then left clicking on the field to edit.

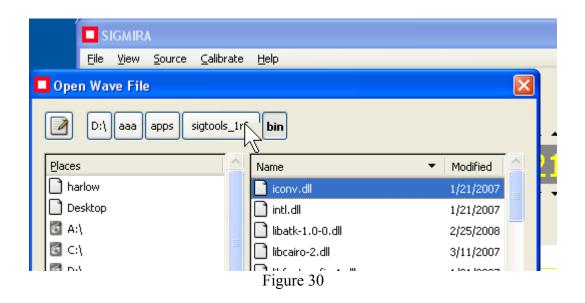
Entries with the gray background can be saved to the local sigmira.ndb file. Select File->Save. Entries from the online database (yellow background) are not saved to the local file.

7.4 Network Receive Disconnect

The Disconnect button can be used to disconnect from a receiver. When Sigmira is currently connected to networked receiver and it is desired to connect to a different receiver or to switch to a different input source (e.g. a local SDR-IQ) it is not necessary to click the Disconnect button first. Sigmira does the disconnect automatically.

8 Wave file Input

Sigmira can operate on prerecorded .wav files without an external receiver connected. Start the program and from the top menus select: File => Open. To find the example files first click on the "Sigtools" button:



The "examples" directory will then appear. Double-click on it:

Open Wave File		×		
D:\ aaa apps sigtools_1r2 bin				
Places	Name 🔻	Modified		
harlow	🗁 bin	Yesterday		
Desktop	🗁 doc	Monday		
🔄 A:\	🗁 etc	4/11/2008		
🖸 C:	🗁 examples 📉	4/11/2008		
🔄 D:\	D lib 7	4/11/2008		
🔄 E:\	🗁 share	4/11/2008		
🔄 F:\	readme_sigtools_1r2.txt	4/6/2008		
🖾 G:\ 🗸	🗋 uninstall.exe	4/11/2008		
Add Remove				
		<u>O</u> pen		

Figure 31

Then select the RTTY example file by double-clicking on it:

Open Wave File			
D:\ aaa apps sigtools_1r2 examples			
Places	Name	•	Modified
harlow	am_7r355_070611_0710.wav		6/12/2007
Desktop	am_7r375_070611_0645.wav		6/12/2007
🖸 A:\	🗋 cw_070114_0631.wav		1/20/2007
🖾 C:	pskfest_070113_1130.wav		1/20/2007
🔄 D:\	📄 rtty_070113_1153.wav 📉		1/20/2007
🔂 E:\	7		
🔄 F:\			
🔄 G:\			
<u>A</u> dd <u>R</u> emove			
	Cancel		Open

Figure 32

The example file will then start playing. Click on the Spectrum Width 5 kHz button. Right-click on the 1070 mark on the waterfall frequency scale. The display will look something like:

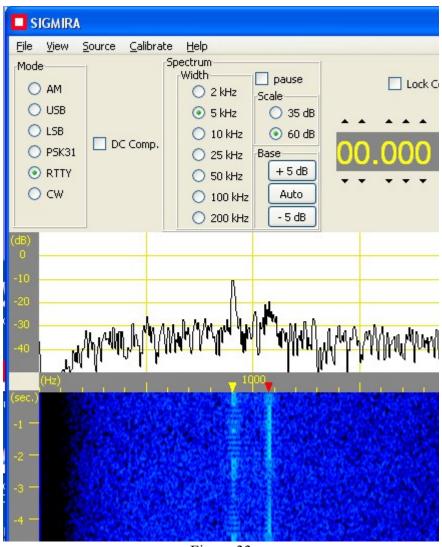


Figure 33

Click on the RTTY mode 'radio button' as described in section 3.5.

The demodulated message will then start to appear:

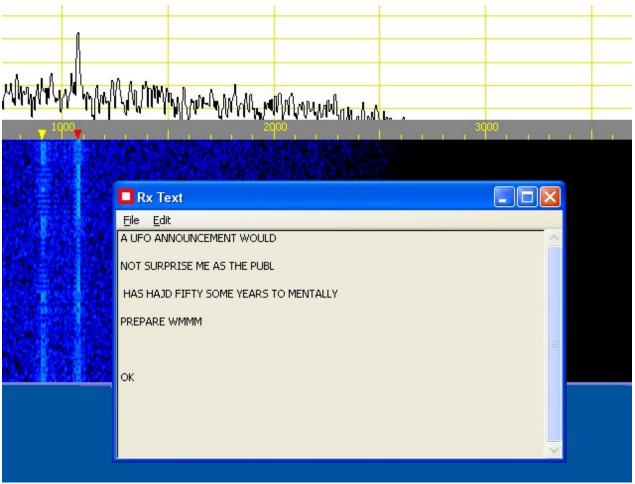


Figure - 35

9 Credits

Sigmira[™] is Copyright 2021 by Steven A. Harlow. All rights reserved. http://www.saharlow.com/technology/sigmira Sigmira[™] is a trademark of Steven A. Harlow.

SDR-IQ[™] and SDR-14[™] are trademarks of RFSpace, Inc. <u>http://www.rfspace.com/Home.html</u>