

# OpenHPSDR for VHF/UHF/Microwave, version 3.0

## by Roger Rehr W3SZ

**I. Introduction** The first presentation that I gave on the use of DSP techniques in Amateur Radio was fifteen years ago, in 2001. At that time I presented at the NEWS Conference a talk on the use of DSP techniques in Amateur weak signal work, titled “A Brief Discussion of Some Software DSP Solutions I've Tried”. I did not use the term “Software Defined Radio” once. The software I discussed included Brian Beezley K6STI's DSP Blaster, Leif Asbrink SM5BSZ's MS-DOS PC-Receiver, which Leif was just porting to Linux at that time, and which had not yet received the name “Linrad”, Bob Larkin W7PUA's DSP-10 hardware and software, and Spectran, by I2PHD and IK2CZL. Gerald Youngblood's now famous 4-part QEX series titled “A Software-Defined Radio for the Masses” had not yet appeared, but would be published in the second half of 2002 and early 2003. The first Flex Radio product, the SDR-1000, would appear in 2003, two years after that talk.

A lot has changed since then! Software defined radios are now used by experimenter and appliance operator alike. There are SDRs with knobs (e.g. Elecraft KX3) and SDRs without knobs. There are SDRs usable by the appliance operator (e.g. Elecraft KX3, FlexRadios) and those more suited for the experimenter (openHPSDR projects). Although I was able to cover the totality of the DSP solutions available to the Amateur Radio operator in 2001 in one paper, that is no longer possible. In this paper I will focus on the openHPSDR project, or High Performance Software Defined Radio project, as these are the radios that I use at W3SZ. The openHPSDR project was originally called just “HPSDR”, but for a variety of reasons not germane to this discussion, the project is now officially termed the “openHPSDR” project.

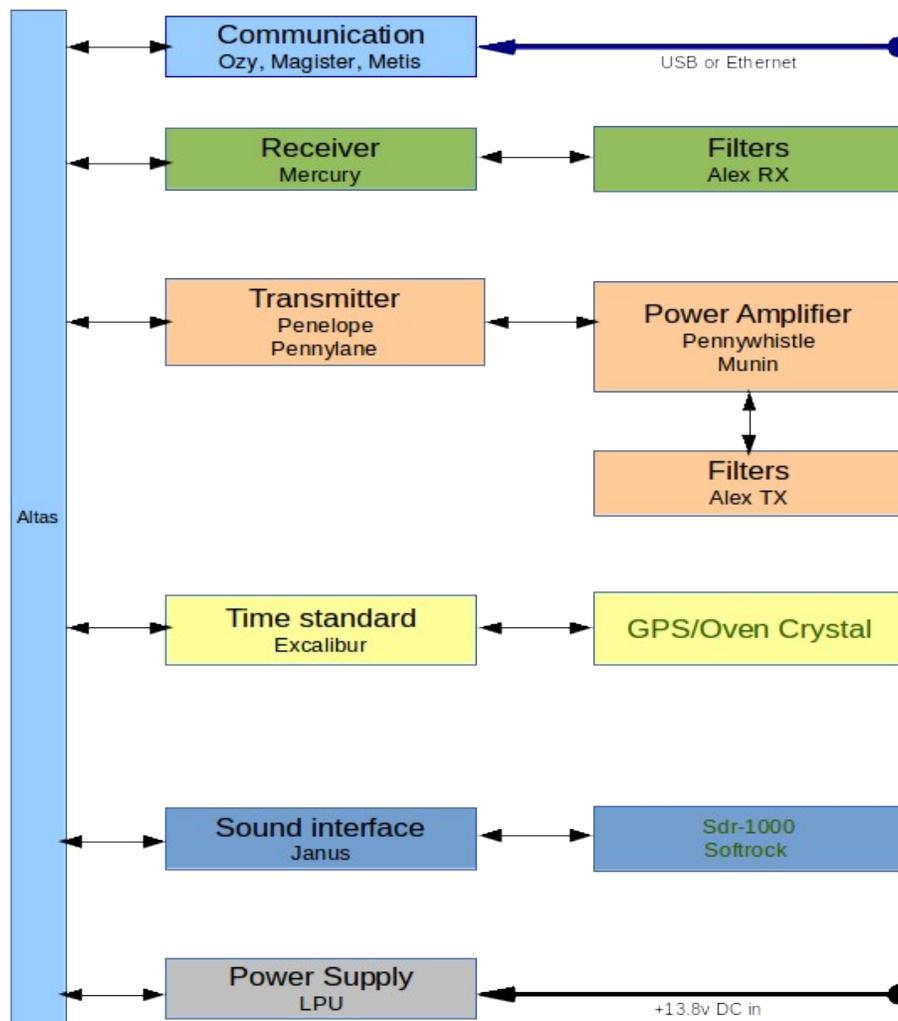
I have previously presented talks and papers detailing the great importance of having full-time bandscope for VHF/UHF/Microwave work (including a talk in 2012, where I presented the first iteration of my openHPSDR VHF/UHF/microwave controller), and hopefully everyone reading this is convinced of the necessity of having full-time bandscope on each of the bands 50 through 432 (or 50 through 903 or 50 through 1296 MHz; I won't argue the small point of where the cutoff lies), with an additional bandscope for 2.3 GHz (or 903 MHz or 1296 MHz) and up. The reader is referred to prior presentations [in this paper](#) and [in this paper](#) and [at this link](#) if he/she remains in doubt on this point.

The primary focus of this paper will be the software I wrote to provide for a transceiver system with multiple simultaneous bandscope for 50, 144, 222, 432, 903, 1296 MHz along with a simultaneous bandscope shared by the bands 2.3 through 24 GHz. This software allows remote operation as well as onsite operation for 50 MHz through 24 GHz with no need for manual bandswitching of antennas, transverters, microphones, headphones, keyers, footswitches, or any other apparatus. The software is fully integrated with N1MM+ and provides for digital audio and COM port connections with N1MM+ and WSJT.WSJT-X etc. I have used this software on a daily basis for remote operation of my microwave station at frequencies from 50 MHz through 24 GHz, and also for on-site contest operation.

**II. History of openHPSDR** In 2005, Phil Covington started the High Performance SDR (HPSDR) project, using a motherboard with an FPGA and a USB 2.0 interface. Around the same time, Phil Harman (then VK6APH and now VK6PH) and Bill Tracey, KD5TFD developed a sound card replacement for the SDR-1000, also using a motherboard with an FPGA and a USB 2.0 interface. This was the XYLO SDR group. These groups merged in early 2006, and HPSDR.org was created. Phil Covington developed the ATLAS backplane and the OZY board, to replace the XYLO board. He then developed the Quick Silver board, which was the initial prototype for the Mercury receiver board.

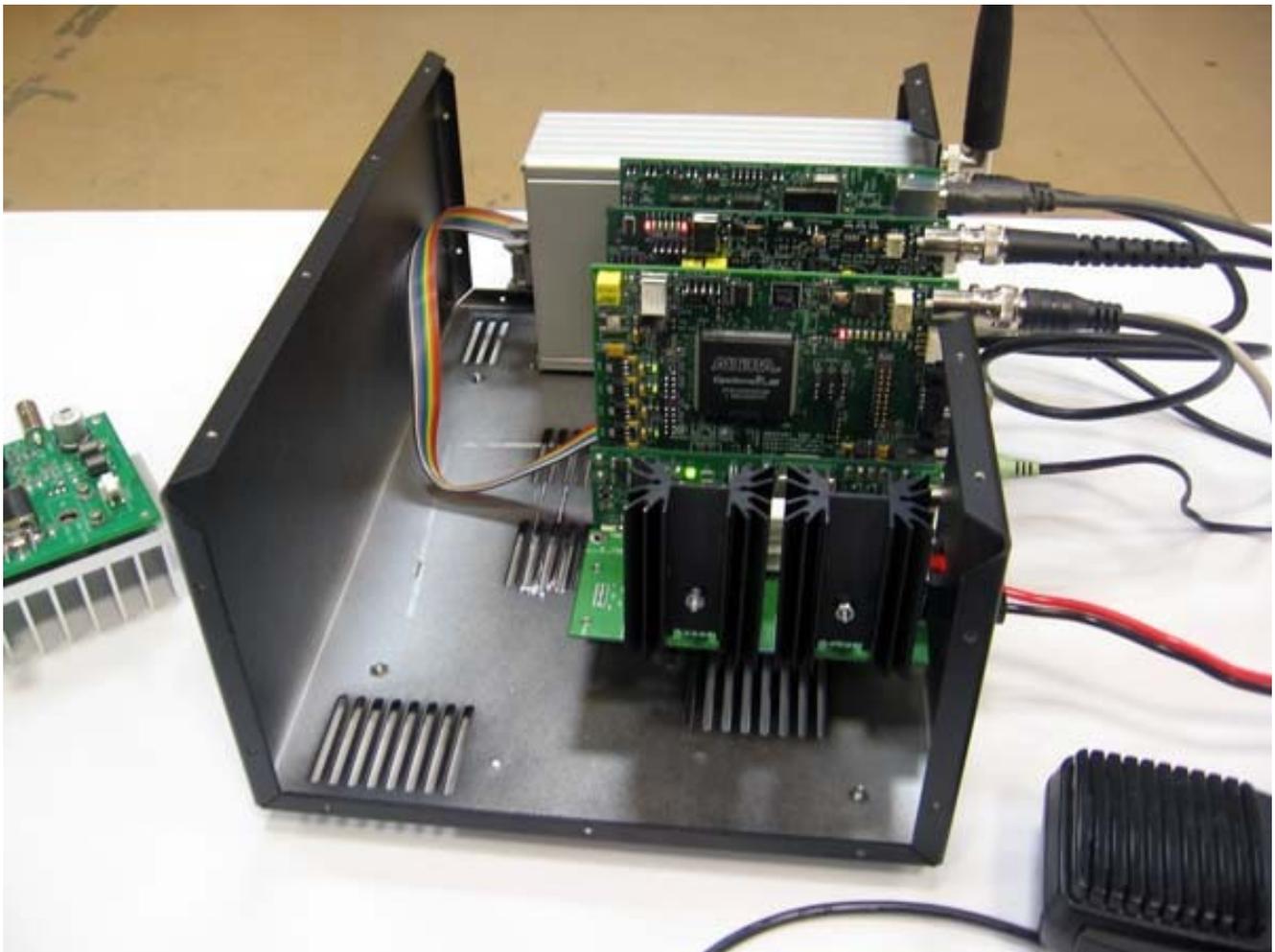
Mercury development began in 2006 and continued through 2007, with Phil Harman VK6PH acting as project leader. In March, 2008 TAPR agreed to fund the Mercury development proposal. Since that time TAPR has continued to provide development funding for the openHPSDR hardware, and the openHPSDR hardware has been available for purchase on the TAPR website. Additional sources of the openHPSDR hardware have included Gerd Loch of Loch Leiterplatten GmbH, and more recently, Apache Labs, owned and managed by “Abhi”, Abhishek Arunoday Prakash, who was the primary PCB layout team member for Hermes. I recommend that you peruse both the TAPR website at [http://www.tapr.org/hpsdr\\_index.html](http://www.tapr.org/hpsdr_index.html) and the Apache Labs website at <https://apache-labs.com/> if you are interested in purchasing openHPSDR hardware. To get a more complete view of the openHPSDR project than I can provide here, refer to the openHPSDR website at <http://openhpsdr.org/>

**III. First Generation Architecture** The initial openHPSDR architecture was that of a common backplane into which various plug-in boards were connected. The graphic below, taken from the openHPSDR website (openhpsdr.org), gives further detail:



In the graphic above you can see the Atlas backplane running vertically on the left of the graphic. Atlas can accommodate up to 6 daughter boards. At the top, connected to Atlas, is a communication board, connecting the openHPSDR hardware to a computer. Ozy was the first such board, and it connected to the computer by USB. Magister was an updated version of Ozy. Metis is the most recently released communications board, connecting the openHPSDR hardware to the computer via Ethernet. Moving down the Atlas bus, we see next Mercury, the Direct DownConversion (DDC) receiver. Alex optionally connects to Mercury to provide filters. I do not use Alex for my VHF and up work, as it is not necessary for this purpose. Next in line is a transmitter board. Penelope was the first Direct UpConversion (DUC) transmitter produced by this project; Pennylane is a newer version. Either of these transmitters optionally connects to an amplifier and the Alex filters. Next in line is Excalibur, which is an optional time/frequency standard board, with optional external 10 MHz input for GPS locking. Janus was a sound interface card, which is no longer needed. Finally, LPU is a linear power supply that takes 13.8 VDC in and supplies appropriate voltages to the Atlas-connected boards. An enclosure box, named Pandora, is available for this system. Pandora measures approximately 12 x 9 x 5.5 inches.

My current station uses this Atlas backplane model for some of its radios, as well as some Hermes radios that I will discuss below. Below is a picture of Pandora with 4 installed daughter cards. I took this image from the openHPSDR website, as I was too lazy to open up one of my Pandora boxes to show the insides.

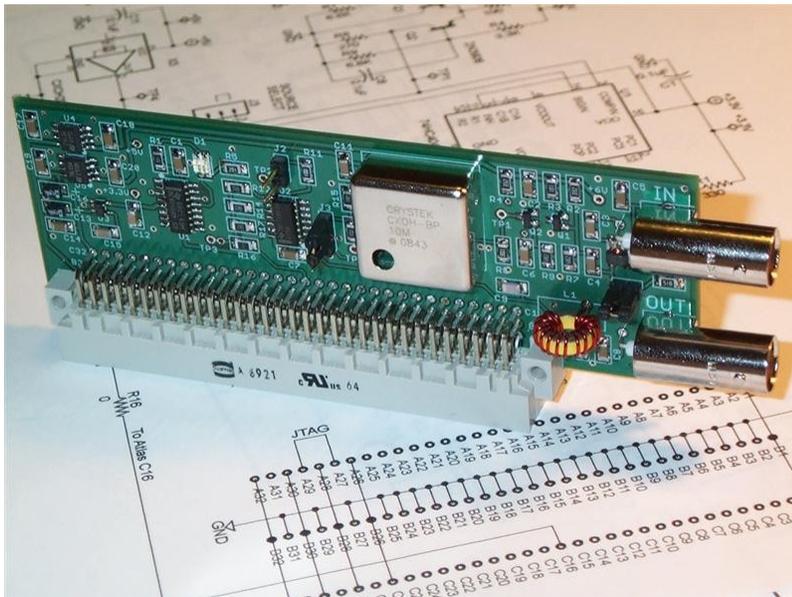


The board in the foreground is the LPU power supply. The next board is the DDC receiver Mercury, the board above that is the DUC transmitter Penelope, the board above that is the USB computer interface board Ozy, and the hardware enclosed in the aluminum case at the very top of the picture with ribbon cable coming out the back is Alex, the set of HF filters. The board sitting behind the Pandora with a large heat sink visible below the circuit board is the PennyWhistle 160-6 meter amplifier, which will deliver 16-20 watts output with 250 mW drive supplied by either Penelope or Pennylane.

Here is a picture of the Atlas backplane with its 6 board slots:



The Excalibur GPS-disciplined 10 MHz source is below:



The Mercury DDC Receiver is shown below:

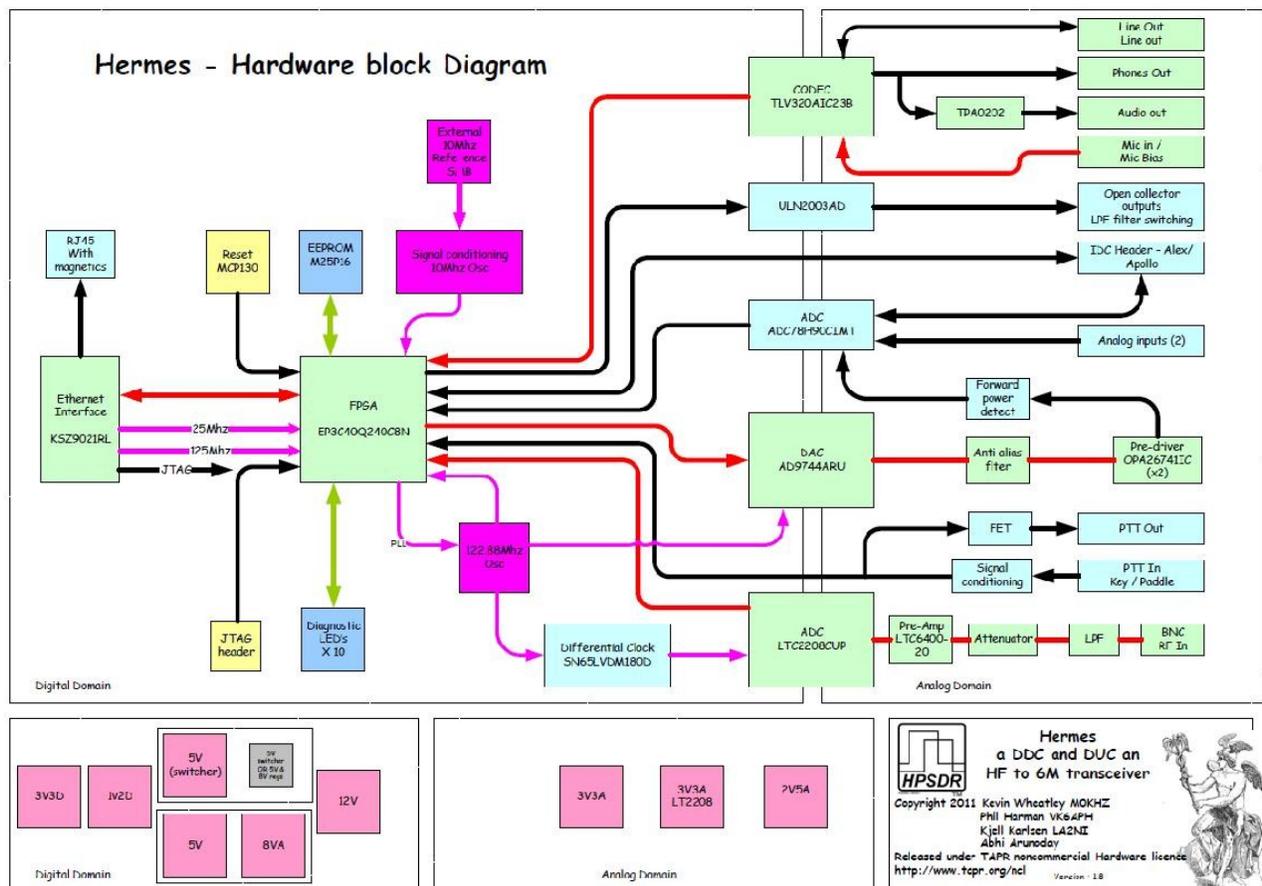


Below is the Penelope DUC transmitter.



**IV. Second Generation Architecture** In 2009 it was decided that a single board transceiver would be the next openHPSDR project. It was to be named Hermes. The Mercury transmitter and Penelope receiver code were successfully placed into one FPGA in 2009, and in 2010 it was decided to add an Ethernet interface to the board. By mid 2012 the Hermes boards were available for purchase on the TAPR website, and on October 1, 2012 Hermes became available from the Apache Labs website as well.

A block diagram of Hermes is below, taken from the [openHPSDR Hermes Wiki page](#):



Both Hermes and the earlier Atlas/Penelope/Mercury hardware provide transceive capability covering approximately 0-55 MHz. Published performance Specifications include:

Blocking Dynamic Range: no detectable gain compression below ADC overload

Dynamic Range 125 dB

Image Rejection > 110 dB

Full Duplex

Transmitter two-tone 3<sup>rd</sup> order IMD -50 dBc @ 400 mW output

500 mW RF output on 160 – 10 m amateur bands, 350 mW on 6 m

Noise Floor -135 dBm in 500 Hz

Seven user-configurable open collector outputs

Separate open collector PTT connection

Stereo outputs at line level and headphone level

Low phase noise master clock (-140 dBc/Hz @ 1 kHz at 14 MHz)

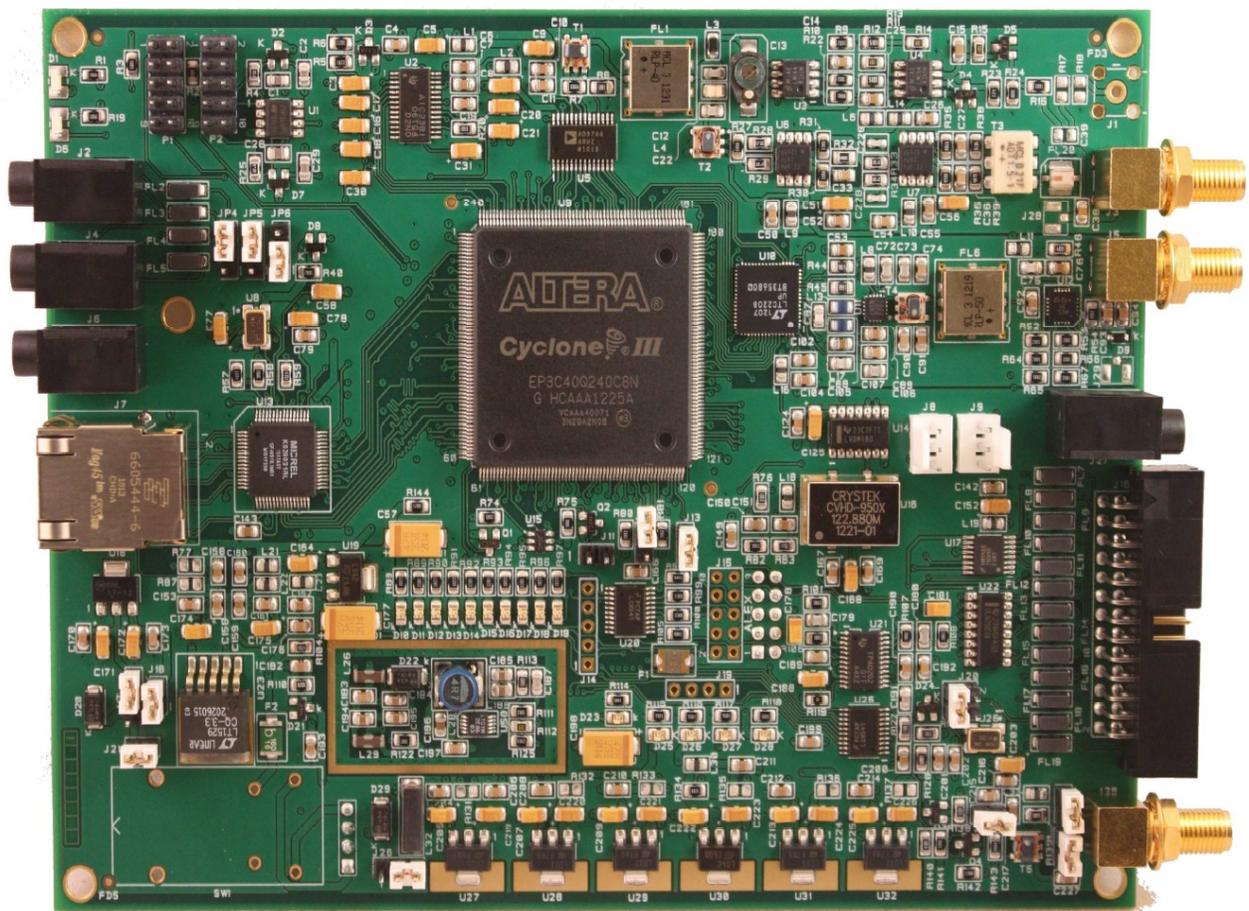
Hermes board size 160 mm x 120 mm (8 layers)

Below is a picture of the Hermes board, taken from the [Hermes User Manual, version 1.18](#).

On the right edge near the top are two gold SMA connectors, used for RF out and RF in. Below that is the 1 Watt audio amplifier output. Below that is the edge connector with 26 pins for PTT, line level stereo audio in and out, stereo headphone out, 3 digital inputs, 2 analog inputs, 7 open collector outputs, speaker output, and several ground connections. The gold SMA connector at the bottom of the right edge is for 10 MHz in, in case GPS locking is desired.

On the left edge, the top connector is for microphone in. The next connector is for headphone, and the third and last connector in that triad is for CW key/keyer.

Below that on the left edge is the RJ45 Ethernet connector.



Below is a photo of a Hermes board that I've placed in an ANAN-10 case, sitting on top of a Pandora box, so you can get some idea of the relative sizes of these two units.



**V. Apache Labs** As was noted above, “Abhi”, Abhishek Arunoday Prakash, founded and owns and manages [Apache Labs](#). Apache's current openHPSDR-related products include the ANAN-10E, the ANAN-10, the ANAN-100, the ANAN-100B, the ANAN-100D, the ANAN-200D, and the Angelia. All of these radios have receive coverage from 10 kHz through 55 MHz and transmit coverage of the Amateur bands from 160 through 6 meters. The Angelia has a maximum RF output of 500 mW. The ANAN-10E and the ANAN-10 have a maximum RF output of 10 Watts, and the other ANAN radios have a maximum RF output of 100 watts. The Angelia is a special purpose product, with dual phase-locked receive channels suitable for diversity reception. The ANAN-10E and the ANAN-100 permit up to 4 receivers, and the ANAN-10, ANAN-100D and the ANAN-200D permit up to 7 receivers. The ANAN-10E and the ANAN-100B use a 14 bit ADC, while the other radios all use 16 bit ADCs. The ANAN-10E and the ANAN-100B use an EP3C25 FPGA. The 10 and the 100 use the larger EP3C40 FPGA, whereas the 100D uses an even larger EP4CE115, and the 200D uses the still-larger EP4CGX150. The 10 and the 100 have 16 MB of FLASH RAM, and the 100D and 200D have 128 MB. The 100D and the 200D each have 32 Mbit of SRAM; the 10 and the 100 have none. A spreadsheet comparing several of the different ANAN models is [here](#). Apache Labs also [sells the](#)

[Hermes board](#), which is no longer available from TAPR.

**VI. Hermes-Lite** There is currently an openHPSDR interest group led by Steve Haynal, KF7O that is working on a cheaper version of Hermes, called Hermes-Lite. The main differences between the units are [1] Hermes-Lite uses a 12 bit ADC rather than a 16 bit ADC, thus reducing digital dynamic range, [2] Hermes-Lite uses a BeMicro SDK FPGA board instead of an onboard FPGA, [3] the Hermes-Lite will cover 0-30 MHz rather than 0-55 MHz. Anticipated cost for those who build their own is \$150 or less. There is a GitHub project page for Hermes-Lite at <https://github.com/softerhardware/Hermes-Lite>

**VII. Hermes VNA.** Both Phil Harman (VK6PH) and Alex Shovkoplyas (VE3NEA) have written software that will turn the Hermes transceiver into a Vector Network Analyzer (VNA). An executable for Phil's software is available on the [openHPSDR downloads page](#). The source code is available [here](#), and as of the time of this writing the executable that is obtained by compiling this source code is newer and has better function than the executable file on the openHPSDR page. Alex's VNA software is called Ham VNA and is available [here](#).

**VII. Receiver and Transceiver Software.** The most full-featured and most widely used software for the openHPSDR radios is PowerSDR, which started with source code modified primarily by Bill Tracey (KD5TFD) and Doug Wigley (W5WC) from the FlexRadio PowerSDR software. PowerSDR is written using Microsoft's C# (C Sharp) programming language, and runs on Windows. Power SDR has been very actively maintained by Doug, with Warren Pratt (NR0V) making major contributions including replacing the original DttSP DSP code with his wdsp library and adding a new and novel noise blanker to name just a couple of his many contributions. PowerSDR (and the other software discussed in this section) can be downloaded from the [openHPSDR downloads page](#).

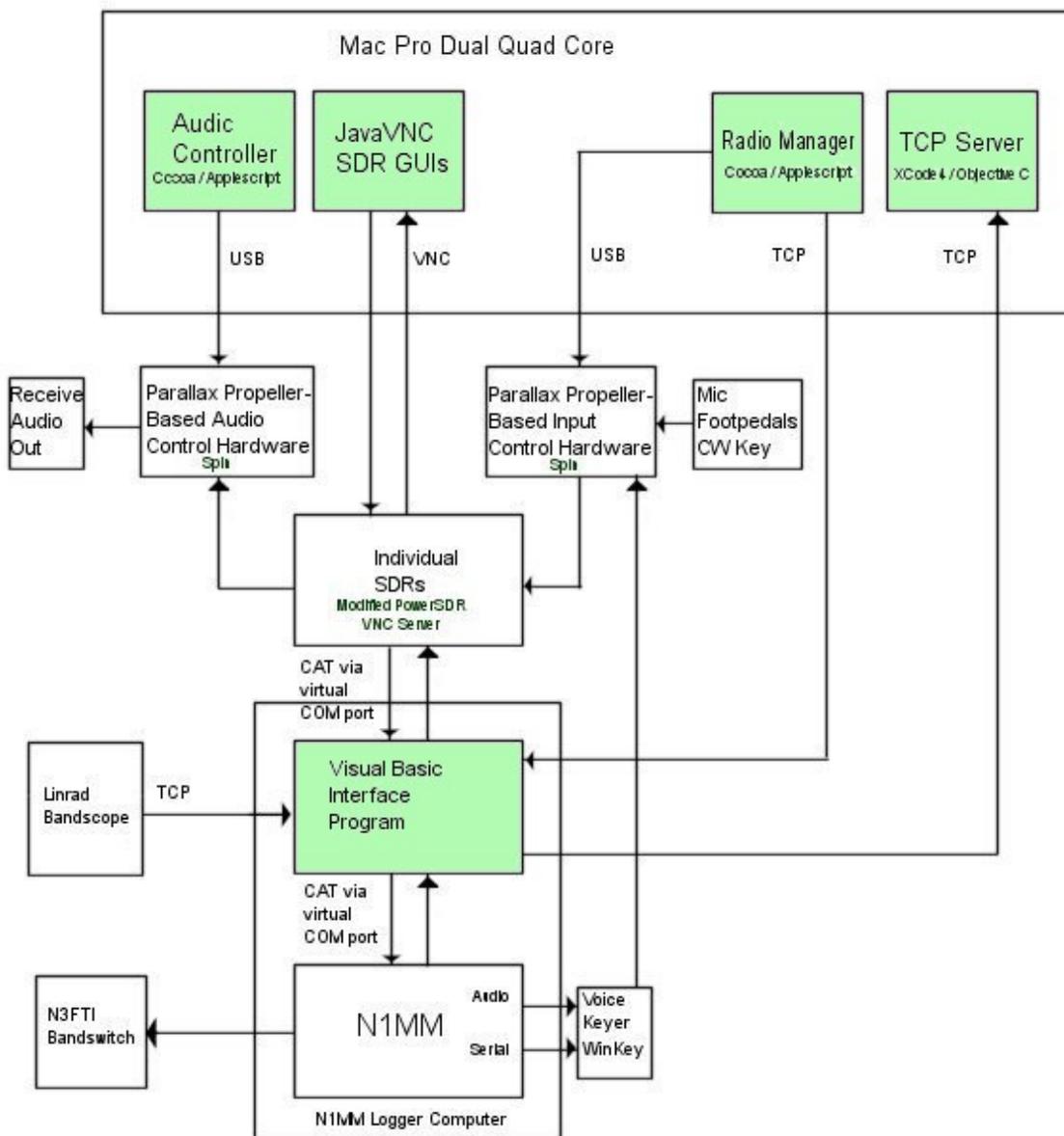
There is also cuSDR, which requires NVIDIA CUDA hardware and, as of the time of the writing of this paper, remains receive only. Like PowerSDR, cuSDR is Windows-based. GHPSDR and GNURADIO-HPSDR are other available software packages for the openHPSDR hardware, and both provide transceive capability.

The final piece of generally available software that we will discuss here is KISS Konsole, which also provides transceive capability. It was developed by Phil Harman (VK6PH) as a simpler alternative to PowerSDR, suitable for experimentation and exploration by HPSDR users. I have used it as the basis for the software that I have developed for my VHF/UHF/Microwave station and which I will discuss below. KISS Konsole is Windows-based and is written in C# (C Sharp), although in mid 2014 Jae Stutzman (K5JAE) reported porting KISS Konsole to Linux, using the [Mono Development Kit](#).

The ability for "users" to play with the code contained in KISS Konsole and PowerSDR was substantially enhanced when Microsoft decided to make the complete Visual Studio IDE (Integrated Development Environment) available as a free download in 2014. Previously, only limited versions of the Visual Studio software had been available at no cost. Currently Visual Studio Community 2015 is the most recent version. You may read about and download Visual Studio Community 2015 [here](#).

If you are serious about programming with Visual Studio, then you should also install NuGet, which is a package manager for the Microsoft development platform. With NuGet you will gain access to many third party addons to Visual Studio. You can download NuGet from [here](#).

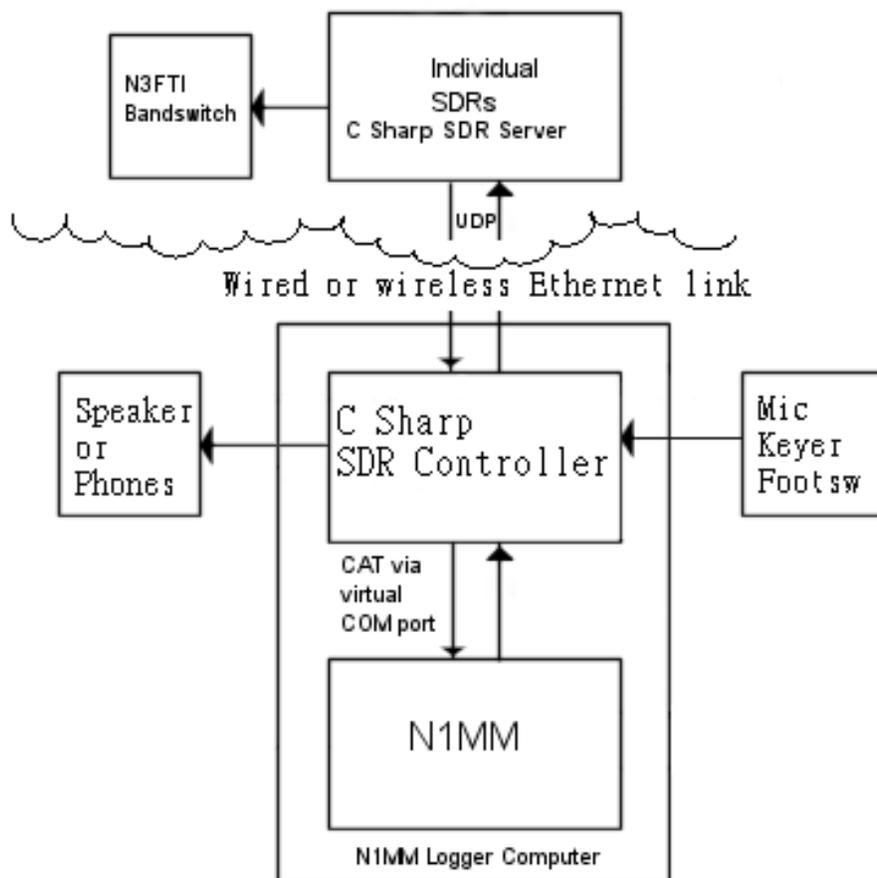
**VIII. My Modified KISS Konsole Software Project.** In 2012 I presented at the combined PackRats/NEWSGroup VHF/UHF/Microwave Conference a paper titled, "What's All This Multiple Bandscope Stuff, Anyhow?". At that time I described the software and hardware that I had built to give me "always on" bandscopes on 50, 144, 222, 432, and 1296 MHz, with an additional always on bandscope that was shared among 903 MHz and 2, 3, 5, 10, and 24 GHz. That system has worked very well for me over the years. The station control GUI actually consisted of several interlocking software pieces, which is a less elegant solution than would be a solution consisting of only a single program. Additionally, because my solution used multiple instances of VNC (Virtual Network Computing), one for each radio, this solution was not as efficient as it might be in terms of network utilization, although I never experienced any network issues, due to the fact that I carefully chose the VNC parameters to minimize network resource utilization. The illustration below, although simplified, gives some idea of the complexity of this earlier solution. Each green-colored box represents a different piece of software written by me for this earlier project:



My primary goals with the current project that I describe in this paper were four. The first goal was to implement the entire Control system as a single program that would seamlessly connect with the N1MM+ logging program, and be seen by N1MM+ as a single radio covering all bands from 50 MHz through 24 GHz. The second goal was to reduce substantially the network bandwidth required for communication between the individual radio/SDR servers and the SDR Controller software. The third goal was to eliminate the need for a Hardware Controller. The fourth goal was to provide the capability for remote operation from a computer not co-located with the radios, by means of either a wired or wireless Ethernet connection.

I had to run my previous system on a Mac Pro Dual Quad Core with Apple's Gigabit network interface, as I needed this horsepower in order to achieve adequate bandwidth for the multiple incoming VNC instances. At the time I put that system together, Apple's Ethernet performance was far better than that available with any easily available Windows-based network interface cards. I was very successful in reducing both the network bandwidth required and CPU utilization with the new software, even though the new software provides for simultaneous streaming of receive audio from two SDRs to the client computer in addition to seven (or 8 if an HF band HF is included) full time bandscopes, radio control, etc. The network bandwidth of the new system is on the order of 0.5% of a 100 Mbps bandwidth connection, that is, on the order of only 500 kbps. Running on a 4.00 GHz Intel I7 4790, the current system has on the order of 45-50% CPU utilization. On a 3.60 GHz Intel I3-4160, the CPU utilization is on the order of 50-55%.

The block diagram of the new system is below.



You can see that the controller now consists of only one program, labeled C Sharp SDR Controller, that interfaces with N1MM + and the individual SDRs and their C Sharp SDR Servers. The link with N1MM is via virtual serial ports, and the link with the radios is via a wired or wireless Ethernet link, using UDP packets. Not shown are WSJT/WSJT-X which also interface digitally with the controller software for exchange of audio and COM port data.

To create the new SDR Controller, I basically wrote two new pieces of software, both derived from the KISS Konsole model. One piece is a server that is directly connected to the openHPSDR radio hardware, and that does the DSP processing and radio control. This server software runs on a "headless" computer with no video monitor, keyboard, or mouse, and the operator has no direct communications with that computer / software. All communications are through the second piece of software, the client. There are no physical connections to the SDR server or its associated radio hardware for microphone audio input, CW key input, PTT input, or receive audio output. All of these connections are made via the Ethernet. The only physical connections to each radio are the Ethernet cable, 13.8V power cable, and the PTT output from the radio for controlling sequencers, linear amplifiers, transverters, etc.

The second piece of software, the client software, is the C Sharp SDR Controller shown in the block diagram above. The C Sharp SDR Controller software is the user interface, and is located on the logging computer at which the operator sits. That computer can be located either at the radio site or remotely, as long as there is an Ethernet connection available. This SDR Controller accepts input from the user, either directly or via N1MM, and it is responsible for all communications with the individual radio/server units, and for assigning footswitch, microphone or other (e.g. N1MM+, WSJT-X, WSJT) transmit audio, CW key/keyer, receive audio, and transverters to the appropriate IF radio automatically with no need for user intervention. The CW key, microphone, and footswitch are all connected to this computer.

With the new software system there is no longer any need for a hardware controller as was used with the old software system, and support for that hardware controller has been dropped from the most recent versions of this software.

To describe the project in list form:

In the first iteration of this software, before I eliminated the need for the Hardware SDR Controller, the basic VK6PH KISS Konsole Software attributes/functions that were retained in my new Server module included:

1. Connects directly to the radio [via Ethernet]
2. Receives baseband receive IQ data from the radio
3. Receives mic input from the radio
4. Receives key input from the radio
5. Receives PTT input from the radio
6. Sends commands to the radio
7. Sends processed receive audio back to the radio
8. Sends transmit IQ data back to radio

However, with the elimination of the need for the Hardware SDR Controller, the functions of the server change substantially (only 1 and 2 are unchanged):

1. Connects directly to the radio [via Ethernet]
2. Receives baseband receive IQ data from the radio
3. Receives mic or other transmit audio input via the Ethernet from the client and sends appropriate IQ data to the radio
4. Receives key input via the Ethernet from the client and sends appropriate IQ data to the radio
5. Receives PTT input via the Ethernet from the client and forwards to the radio
6. Receives commands from the client and forwards these commands to the radio if appropriate
7. Sends processed receive audio to the client
8. Microwave server receives band information from client and sends band information to N3FTI bandswitch

Other major changes/additions that I made to KISS Konsole when I created the Server module were:

1. Adds capability for server to receive commands from client software, so that server and radio can be controlled remotely by the client software.
  2. Adds adjustable FFT Size for Spectrum and Waterfall. There was previously only one small FFT size that was "hardwired in". This is not acceptable for weak signal work. FFT size is now adjustable from 4096 to 524288. To do this I had to split the audio FFT processing from the graphic FFT processing, and I used [FFTW](#) and the C# wrapper for it named [FFTWSharp](#). I had to extend the FFTSharp dll to implement Wisdom. That is described near the bottom of [this page](#).
  3. Adds CW sidetone using a selectable computer sound playback device. Because C# does not have audio support, I used the [CSCore](#) C# wrapper for Windows audio functions for this.
  4. Adds receive audio using a selectable computer sound playback device. Previously there was receive audio available only at the openHPSDR hardware. Although receive audio was created in the computer from IQ data, it was sent back to the radio and could not previously be listened to through the computer. Because C# does not have audio support, I used the [CSCore](#) C# wrapper for Windows audio functions for this.
  5. Server sends Spectrum/Waterfall data to client over the Ethernet
  6. Server can send CW sidetone data to client over the Ethernet. This is not used with current mode of operation, where CW is generated at the client.
  7. Adds "Wisdom" FFT optimization to server. When server software is first run on a particular machine, optimization of the FFT routines is performed.
  8. Adds zoom of spectrum and waterfall displays.
  9. Adds frequency stepping using up/down arrows on keyboard.
  10. Adds keyboard-adjustable step size for frequency adjustment.
  11. Adds keyboard-adjustable mode selection.
- (9-11 were added so that a [ShuttlePRO](#) could be used to modify these parameters)

The client module has the following main features:

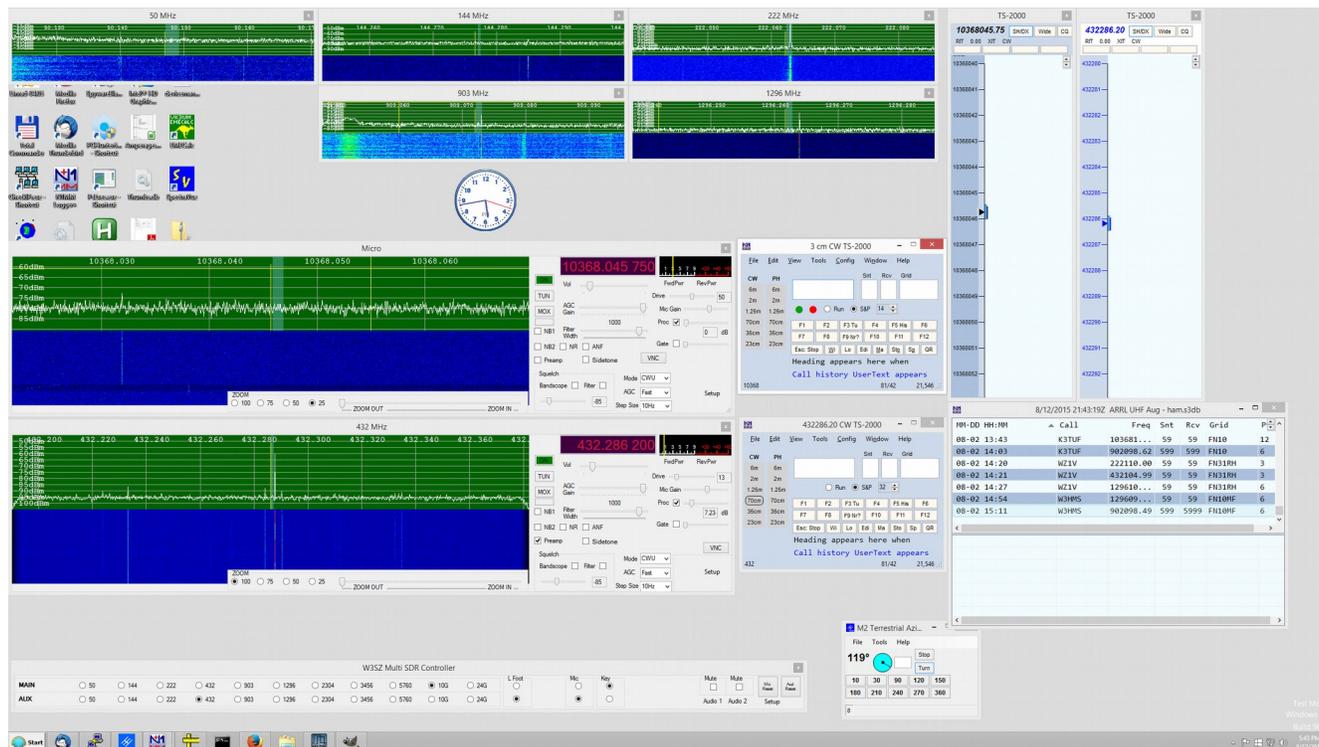
1. Sends via the Ethernet commands to the servers for the radios selected as the Main and Auxiliary radios, to control all necessary functions of these two radios and their associated servers.
2. Constantly receives via the Ethernet and displays Spectrum/Waterfall data from all radios, simultaneously, so that all bandscope are always visible for all bands.
3. Receives via the Ethernet receive audio from the servers for the radios selected as the Main and Auxiliary radios.
3. Interfaces with N1MM+ logging program, appearing to N1MM+ as two Kenwood TS-2000s covering HF bands and all bands from 50 MHz through 24 GHz, inclusive.
4. Connects/switches footswitches, CW key/keyer, microphone or other transmit audio, 2 receive audio channels with appropriate server/radio, automatically,

5. Creates CW sidetone on a selectable audio playback device on logging computer. Because C# does not have audio support, I used the [CSCore](#) C# wrapper for Windows audio functions for this.
6. Adds multiple waterfall palettes (original, enhanced, Spectran, black-white, Linrad, Linrad Log, Linrad Auto)
7. Adds spectrum/waterfall zoom
8. Adds frequency adjustment by up/down arrows.
9. Adds key-adjustable step size for frequency adjustment.
10. Adds key-adjustable mode selection.
11. Adds integrated audio and COM port connections for N1MM+ so that all receive audio for one or both selected radios can be recorded by N1MM+/QSOrder, and so that N1MM+ recorded wave files (CQ, etc.) can be used for transmit audio, with automatic cycling of the selected radio to transmit status when these files are played.
12. Adds integrated audio and COM port connections for WSJT-X, WSJT, etc. so that receive audio from the selected radio can be sent to one of these programs for decoding, and so that transmit audio from the selected program can be used as transmit audio by this software, with automatic cycling of the selected radio to transmit status when appropriate.
13. Provides for audio monitoring of transmit audio, whether it comes from the microphone, N1MM+, or one of the digital programs.  
(8-10 were added so that a [ShuttlePRO](#) could be used to select these parameters)

As described by its creator Phil Harman, VK6PH, "K.I.S.S (Keep It Simple Stupid) Konsole is a straightforward PC program that will allow beginners in SDR and DSP programming to get their feet wet. KK is intended as a learning experience and not as a competitor or replacement for any existing Console code." My use of KISS Konsole as the basis for this project was just what Phil had intended when he first wrote KISS Konsole back in 2009. I took a great starting point and added to it and modified it to make a software system that fulfills my needs for my VHF/UHF/Microwave station.

With this new software, the use of a homebrew hardware controller as I used with the old system to automatically switch two footswitches, the microphone, the CW key, and two receive audio channels to the desired radios is no longer necessary. All switching is automatically done in software, and direct physical connections to the microphone audio, CW key, and PTT input jacks on the SDR radios are not needed and are not made. Because it was previously discussed in detail, and because it is no longer necessary, I will not discuss the hardware controller in detail here. If you wish, you may read more about it [in this paper](#) and [in this paper](#) and [at this link](#).

The image below shows the main monitor screen when the new C Sharp SDR Controller Client software and N1MM are running:



The Server/Client combination provide full-time separate bandscope for 50, 144, 222, 432, 903, and 1296 MHz and another bandscope for 2, 3, 5, 10, and 24 GHz. In addition, an HF bandscope/radio can be enabled if desired. There is full integration with N1MM Plus. As noted above, there is automatic switching of microphone or other transmit audio, CW Key / Keyer, Footswitch, receive audio channels for the two selected radios, and of the 2, 3, 5, 10, and 24 GHz transverters to the microwave IF radio. The system is designed to allow remote operation and I use it daily in this fashion via a 5 mile wireless link between my home and my remote operating site.

On the left side of the screen is the SDR Controller Software Client, and on the right is N1MM+. The small bandscope belong to those bands that are not selected as either the Main Radio or the Auxiliary Radio. The two larger bandscope belong to the Main Radio (on top) and the Auxiliary Radio (below). Starting at top left, the small bandscope in the top left corner is 50 MHz. Next to this small bandscope, on its right, is 144 MHz. Next to that is the bandscope for 222 MHz,. Missing from the first spot in the next row is 432 MHz, which is missing because it has been selected as the Aux Radio. When another band is selected to be the Aux radio, the 432 bandscope will return to small size and to this position. To the right of this radio is the 903 MHz small bandscope. To the right of 903 MHz is the 1296 MHz small bandscope. On the left, below the empty 432 MHz small bandscope space, is another blank space, which is reserved for the Microwave bandscope, which has been selected as the Main Radio. IF an HF radio were selected in the setup menu, to the right of the empty space that is reserved for the small Microwave bandscope the HF small bandscope would appear.

Below the large Main and Aux bandscope is the Controller Bar, which will be discussed more below.

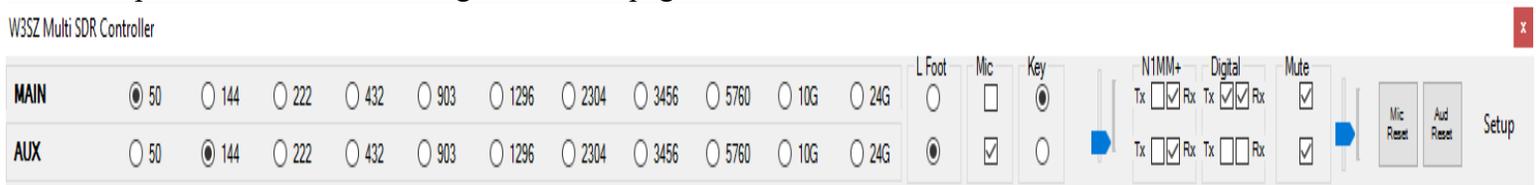
The N1MM+ entry windows for the Main and Aux Radios are situated immediately next to their

respective large radio bandscope. To the right of the N1MM+ entry windows are the N1MM+ bandmaps for these bands. Note that the bandmaps display the same frequency as their respective radio's bandscope, as they should. Below the bandmaps is the N1MM+ log entry list, and below that is the N1MM+ rotor control.

One can change bands in several ways. First, left clicking on a small bandscope will move that band to the large Main Radio position, and return the radio that was in the Main Radio position to its reserved small bandscope position. Right clicking on a small bandscope will move that band to the large Aux Radio position, and return the radio that was in that position to its reserved small bandscope position. Double left clicking on the Main Radio or the Aux Radio while holding down the control key will return that radio to its small bandscope reserved position.

Secondly, one can change bands (or frequencies within a band) by typing the frequency in the N1MM+ entry window. If one types 5760125 into the N1MM+ entry window for the main radio, then the Microwave radio will be placed in the Large Main Radio position and set to 5760.125. Similarly, if one types 432100 into the N1MM+ entry window for the Aux Radio, then the 432 MHz radio will be placed in the Large Aux Radio position and set to 432.100.

Thirdly, one can change bands by using the Controller Bar that is situated below the large Aux Radio bandscope. This Controller Bar is shown below. It appears a bit distorted because I have changed the aspect ratio so that this image fits on the page.



If you are reading this on paper, the details of this controller will be difficult to see. If you are reading it from a PDF file, you can zoom in: 150% zoom will allow you to read the text on the controller bar easily, although you will likely need to scroll from left to right to see the whole bar.

The bar has two rows of "radiobuttons", as they are called. The top row is for the Main Radio, and the bottom row is for the Aux Radio. The left two-thirds of the bar contains radio buttons for each of the bands from 50 MHz to 24 GHz. Clicking on any one of these buttons in the top row will place that radio/band in the Main Radio position. Clicking any one of these buttons in the bottom row will place that radio/band in the Aux Radio position.

Just to the right of the radio buttons are two rows of buttons for the footswitch, the microphone, and the CW key. Clicking on any one of these buttons in the upper row will connect that button's hardware to the Main Radio. Clicking on any one of these buttons in the bottom row will connect that button's hardware to the Aux Radio. Each time a new band/radio is selected as either the Main or Aux Radio, the hardware assigned to each button will be connected to the appropriate radio. To the right of the "Key" Radiobuttons are additional controls. The leftmost vertical trackbar sets the audio level for the transmit audio monitor. To the right of this trackbar is a GroupBox labeled "N1MM+". Checking either the Main or the Aux "Tx" box will send N1MM-generated transmit audio to the respective radio. Checking either the Main or the Aux (or both) Rx box will send the receive audio from the respective radio[s] to the N1MM+ addon QSOOrder. In this manner all contest audio or specific QSO audio or both

can be recorded to disk. To the right of the N1MM GroupBox is the Digital GroupBox. The checkboxes serve similar purposes to those described for N1MM, but in this case they send received audio from the selected radio to a digital program such as WSJT-X, WSJT, etc. and they receive transmit audio from WSJT-X or WSJT, etc. and send it to the selected radio. To the right of the Digital GroupBox is a GroupBox labeled "Mute". Clicking the appropriate checkbox in this GroupBox will mute the speaker/headphone received audio for the selected radio[s]. It will have no effect on audio being sent to N1MM+ or to WSJT-X, WSJT, etc. To the right of the Mute GroupBox is a vertical trackbar for adjusting the audio level of the Main and Aux radios. Additional volume controls are provided on each radio, but this provides a convenient place to adjust the global received audio level without having to shift focus. In one of the setup windows for this program the bands to be displayed can be selected. If a band has not been selected for display, then it will not appear on the control bar, and its radio GUI will not be displayed. This is why the "HF radio" does not appear on this control bar; it has not been selected in the setup.

At the extreme right of the controller bar are buttons to reset the microphone or receive audio. These are primarily for testing purposes, but can be used at any time to reset these audio functions.

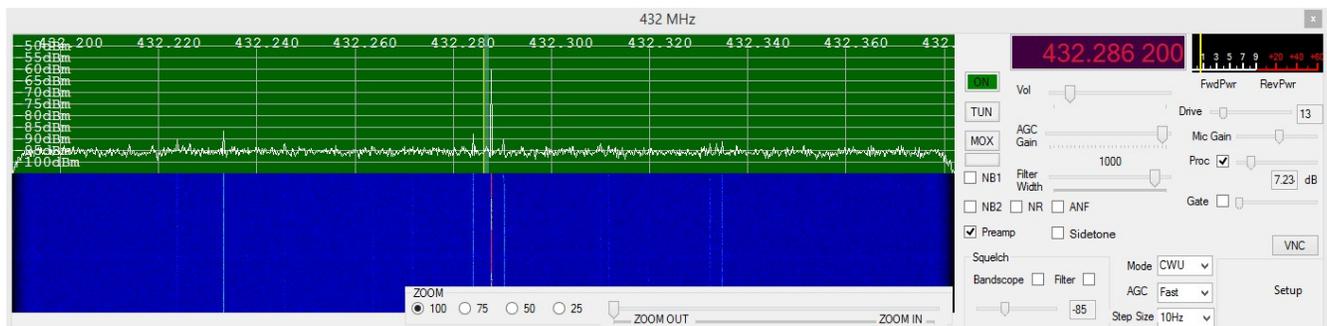
To change frequency within a band, in addition to typing the new desired frequency into the N1MM+ entry window, one can also change frequencies by left-clicking on the spectrum display of the selected radio. The frequency will move to the frequency represented by the point that was clicked on. One can also change the frequency by using the up or down arrows on the keyboard, or by using the knob on a ShuttlePro device if one is available.

The C# SDR Controller communicates with N1MM+ via virtual com ports, using the N1MM+ CAT control facility. N1MM+ thinks that the C# SDR Controller is two Kenwood TS-2000s that cover HF plus all of the amateur bands between 50 MHz and 24 GHz, inclusive.

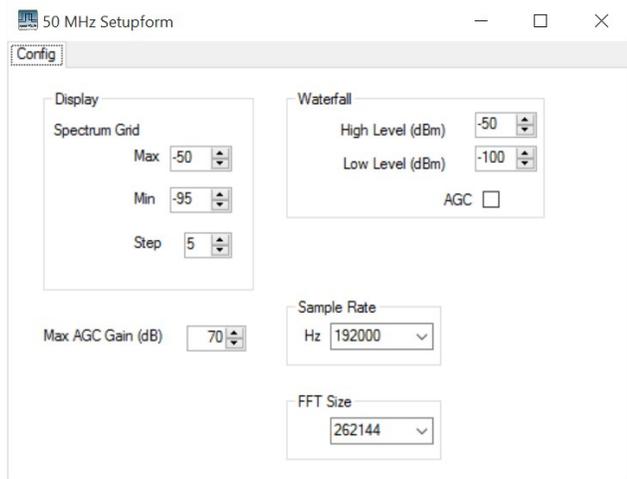
The microwave server controls switching of the transverters for 2 GHz-24GHz via an N3FTI bandswitch. These transverters all use a common IF radio. The lower bands each have dedicated IF radios.

No manual switching of any hardware is required for operation between 50 MHz and 24 GHz. As previously noted, the C Sharp SDR Controller takes care of all hardware switching automatically.

Controls for all radio functions necessary for operation are available on the large Main and Aux Radio bandscopes, as shown in the image below and discussed below. Changing a control value on the large Main or Aux Radio Bandscopes will set the value of that control on the server software, and thus on the appropriate hardware radio. You may want to zoom in on this image to get a better view of the controls.

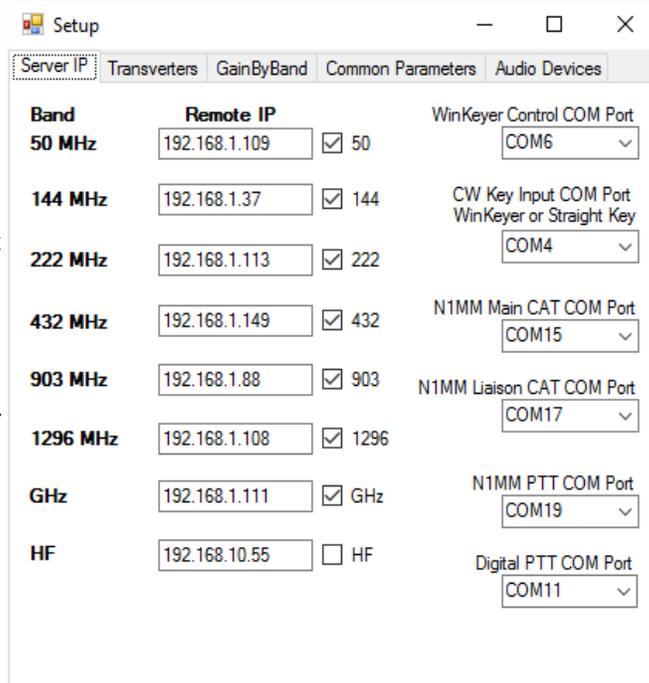


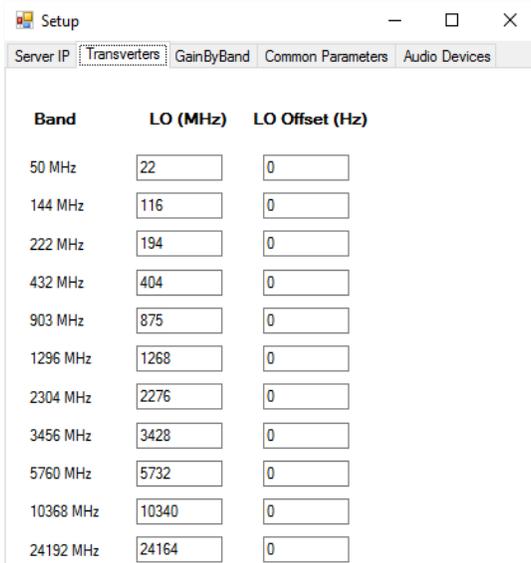
The left three-quarters of the Main and Aux Radio bandscope are occupied by a spectrum display on the top and a waterfall beneath. To these displays I added zoom capability, and also the ability to increase the FFT size up to 524288. I generally use an FFT size of 262144, which gives adequately small bin size (just less than 1 Hz when running a sampling rate of 192 kHz). On the right you can see the digital frequency readout and the S meter at the top, and the "usual" radio controls below including receive volume, AGC Gain, Filter Width, Noise Blankers, Noise Reduction, Automatic Notch Filter, Preamp on/off, CW Sidetone on/off, Squelch controls, Mode, AGC Type, and frequency step size. Transmit controls include Drive level, Mic Gain, Speech Processor on/off and Level, and Noise Gate on/off and Level. In addition there are of course Tune and MOX controls. There is also a Setup button, which brings up a band-specific setup menu to allow the user to set some band-specific parameters from the client. This menu is shown below.



The band-specific setup menu allows the user to set maximum and minimum signal levels for the spectrum and waterfall displays, to activate waterfall AGC if desired, to set the maximum audio AGC level, and to set sampling rate and FFT size.

Commonly used software settings are available from a Setup button near the right edge of the Controller Bar. The five setup menu pages that are accessed via this button are for parameters that are either common to all radios or which are best addressed from a common menu where all bands are available in a group. The first such menu, shown on the right, is labeled, "Server IP". With this page the user [1] sets the IP address for the HPSDR-server for each band (radio), [2] selects the radios that will be displayed, and [3] assigns the COM ports that are used for various functions. Assignment of the IP addresses for each radio server is self-explanatory. The virtual COM ports provide for WinKeyer Control, CW Key/Keyer and Manual PTT input, N1MM+ CAT control for the main and auxiliary radios, and PTT control from N1MM+ as well as from a digital program such as WSJT-X or WSJT.

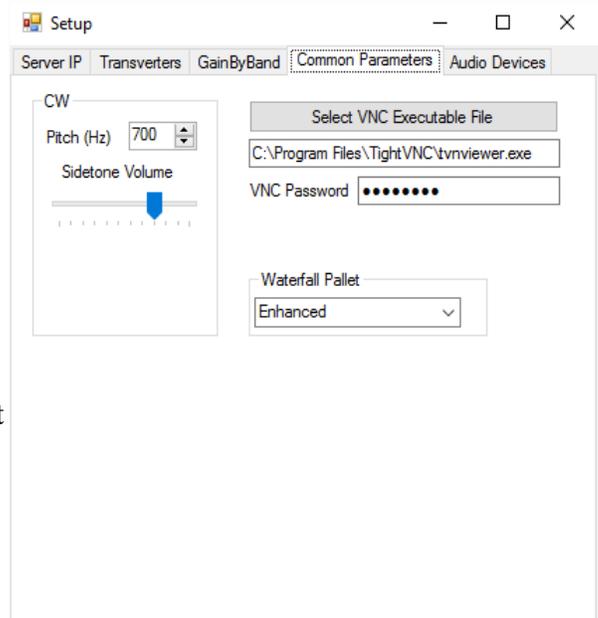




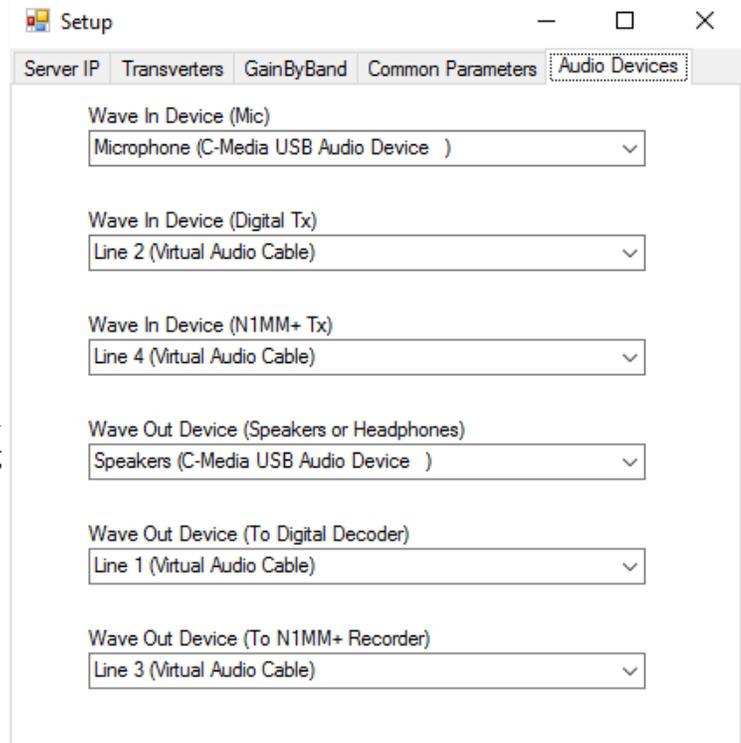
The second setup screen available from the Main Controller Bar of the client software is labeled "Transverters", as shown here. It allows the user to set the LO frequency and offset for each transverter. Because my transverters are GPS-locked on all bands, the LO offset is consistently zero. The third setup screen, not shown here, allows the user to set the transmitter Band Gain for each HF, VHF, UHF, and microwave band. I have a separate, software controlled HP attenuator that is used to set drive levels and receive attenuation levels automatically for each band, to provide optimal transmit drive for each band and a consistent noise floor on receive for each band, and so the third setup screen is not needed in my current setup. It is however maintained in case of future need.

On the right is the fourth menu page accessed by the Setup button on the Controller Bar. This screen is used to set the CW sidetone pitch and volume and to choose the location of the program "Tight VNC" and the VNC password for the remote radio servers. On this page the user can also select the waterfall pallet to be used for all radios.

The VNC button, which you probably noticed sitting just above the Setup button, brings up a VNC connection to the server computer for the selected radio, allowing direct interaction with the server, and providing access to a much larger selection of setup parameters which are of the set once and forget thereafter variety.



The “Audio Devices” setup page allows the user to set the three audio input devices to be used for transmit input for [1] microphone audio, [2] the digital programs such as WSJT/WSJT-X, and [3] N1MM. It also allows the user to choose the three audio output devices to be used for [1] speaker/headphones, [2] input to WSJT/WSJT-X, and [3] input to N1MM+/WSOrder for recording of contest audio. The Windows WASAPI audio system is used for these functions, along with the C Sharp audio wrapper CSCore as noted earlier in this discussion.



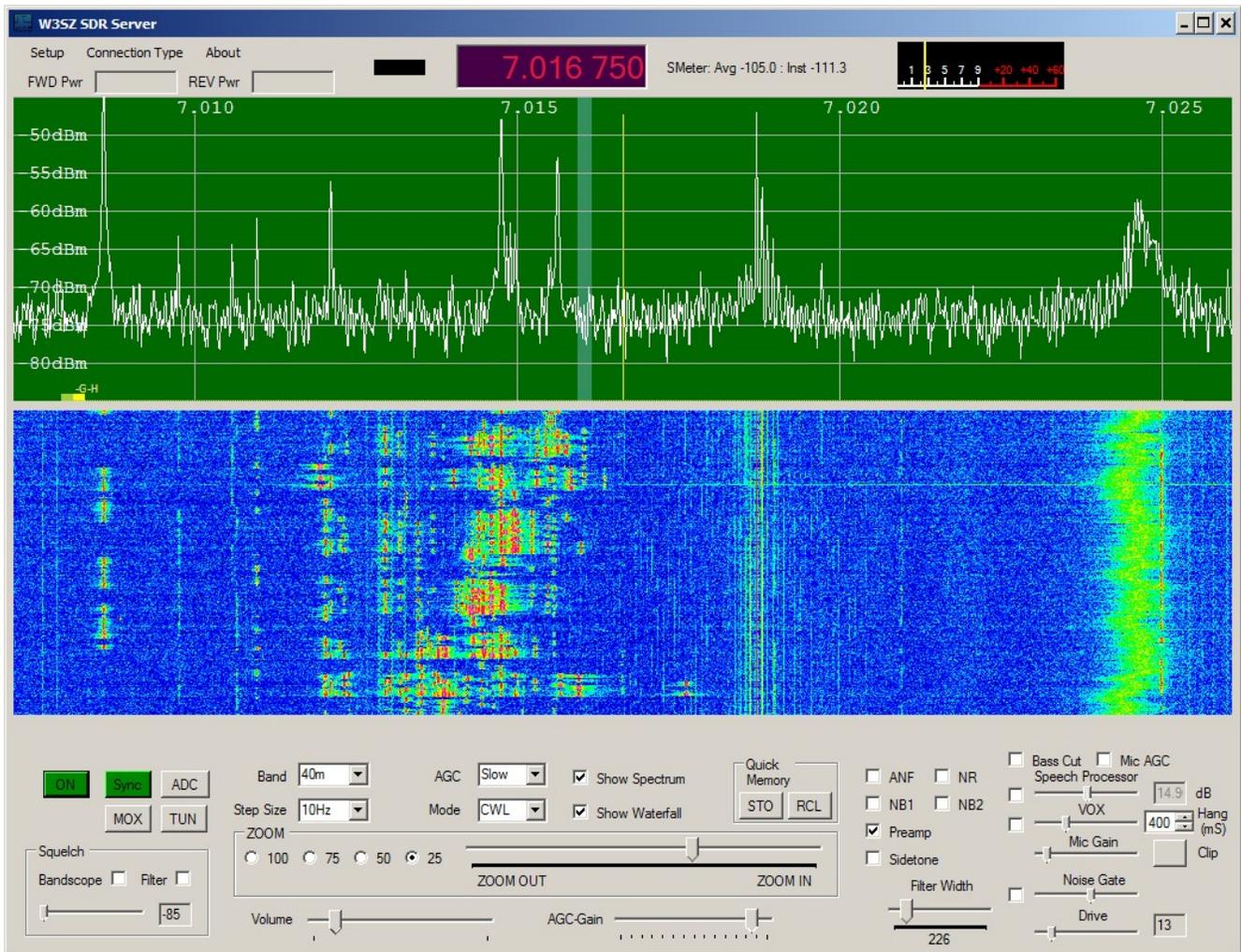
During use, the software system provides a superb operating experience. On one screen the user sees all of the individual bandscope and N1MM+. If the operator sees an interesting signal on one of the small bandscope, he left clicks on that to bring radio to the Main Radio position (or if he wishes, he can instead right click it to bring it to the Aux Radio position). If necessary, he then left-clicks on the signal of interest to center it in the passband. He then just steps on the footswitch (or activates the MOX control on the radio display) and either speaks into the microphone or uses the CW key to send his information to the other station. He types the other station's call and report into the appropriate N1MM+ entry boxes and hits the "Enter" key. If the operator is using the digital modes, he merely clicks on the “Enable Tx” button on WSJT or WSJT-X and operation, except for typing in the station's callsign, is essentially automatic.

Once a station has been logged, the operator is then ready for the next contact. He can either stay on that frequency, or move about that band by either left clicking on the spectrum, using the up/down arrow keys on the keyboard, dialing the knob on the ShuttlePro, or typing a frequency into N1MM+.

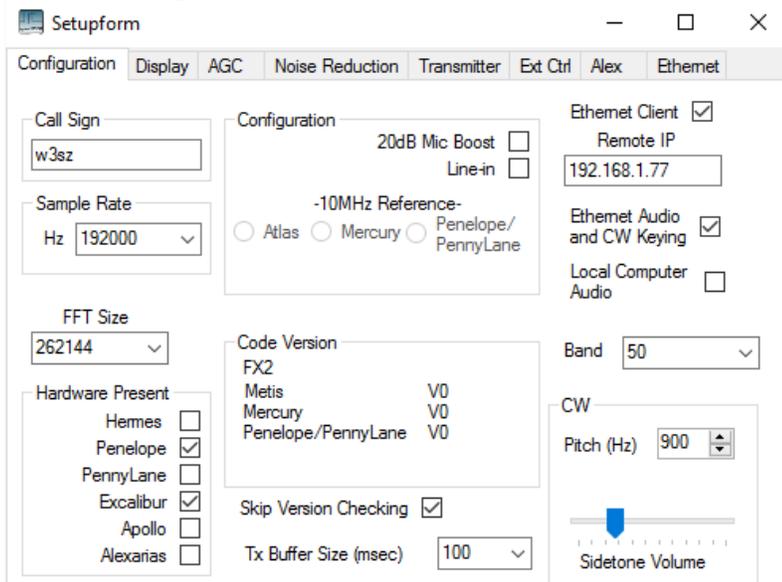
Or, he can change bands by either left or right clicking on the small bandscope he desires to use, which will bring it to the Main or Aux Radio position, or he can click on the radiobutton for the desired band on either the Main or Aux row of the Controller Bar. Or he can type a frequency for another band into the appropriate N1MM+ entry window, and change bands that way, being taken directly to that frequency.

"Running the Bands" from 50 MHz through 24 GHz becomes a matter of just sequentially clicking on the small bandscope for each band (or changing bands by clicking on the RadioButtons on the controller bar, or typing the successive frequencies into the N1MM+ entry window), making the contact, and then repeating the process for the next band. All the while, a second band can be "parked" in the Aux radio position, monitoring the liaison frequency at all times, and ready to transmit with just a tap on the footswitch or a click on the MOX control if it is necessary to establish liaison communications with the other station.

It is not necessary in the course of usual operations to directly access the radio servers, but as noted above the servers can be easily accessed by clicking the VNC button. When that is done, a VNC session of the server immediately pops up on the screen. The server GUI is shown below:



The server's GUI is very similar to, but not identical to, the original GUI from Phil Harman VK6PH's KISS Konsole. One of the major differences is the ability to zoom the waterfall and spectrum, which was not present in the original software. A hint at the wide variety of parameters that can be adjusted from the Setup button on the server's GUI is seen below:



Shown here is the first tab, labeled Configuration. My additions to this tab, that were not present in the original KISS Konsole setup menus, allow the user to set the IP address of the client computer, to specify the frequency band assigned to the server by the client software, to set the FFT size to be used for the bandscope and spectrum display, to set CW sidetone pitch and volume, and to select whether or not to route local audio to the local

computer, to select the buffer size to be used for transmit audio, and whether or not to use ethernet for microphone audio, receive audio, and CW keying.

The original plan when I upgraded this software system was to use the Hardware SDR Controller when I was co-located with the radios, and to use the software-only controller when operating my station remotely over a 5 mile wireless link. However, the software-only controller has worked so well that I am not currently using the hardware controller, even for local operation, and I have removed its functions from the software. A key to the excellent performance of the software-only controller has been keeping Ethernet bandwidth low. Radio control functions and the seven always-on bandscopes consume approximately 450-500 kbps (~0.5 Mbps). When I initially added two receive audio channels with no compression, that pushed the bandwidth up to approximately 2500 kbps (2.5 Mbps). I therefore added the Opus codec to the software. That reduces the required Ethernet bandwidth for the receive audio streams for two radios to less than 50 kbps (0.05 Mbps), so that total Ethernet bandwidth remains on the order of 500 kbps. During transmit mode, the microphone audio channel requires a bandwidth of approximately 25 kbps (0.025 mbps) or less. CW keying requires only a small bandwidth, because I send only state changes from key up to key down or vice versa.

I have placed a short, less than 10 minute, video demonstrating the operation of the system on YouTube. It should be watched in as high definition mode as your system will permit. The URL for this video is: <https://youtu.be/CfOgDMXlpJQ> The capabilities of the software have progressed since this video was made, but it accurately presents the software except for a few additions which were added after the video was made.

**Hardware Connections.** The client uses user-selectable audio devices on the client computer, as shown above. Thus the microphone needs to be set up to be the appropriate Windows recording device and the headphones or speaker need to be set up to be the appropriate Windows playback device. As noted above, the CW key and PTT control (footswitch) are accessed through a COM port on the client computer. As most modern computers do not have legacy serial ports, this is most easily accomplished using a five-dollar USB to serial adapter. Using such an adapter also provides protective isolation between the external hardware and the computer. As designed, the software uses the following pin assignments on the DB9 connector:

Pin 1 (Carrier Detect): Footswitch center pin  
Pin 4 (DTR): CW Key(er) shell connection (common pull-up)  
Pin 5 (Ground): Shield connected only at computer end  
Pin 6 (DSR): CW dash signal  
Pin 7 (RTS): footswitch shell connection (pull-up)  
Pin 8 (CTS): CW dot signal

Pin 1 has a 10K resistor in series with the signal line

**Some Coding Details** The code was compiled with Visual Studio 2013 (and later Visual Studio 2015) and Framework 4.5. A few aspects of the new coding are described below.

1. The large FFT. A large FFT, used to reduce the bin size for the spectrum display and waterfall, is essential for weak signal work. I had previously added large bin sizes to the openHPSDR version of PowerSDR and this was eventually incorporated into the main open HPSDR PowerSDR trunk. I have found that an FFT size of 262144 provides, for me, optimal sensitivity. I included FFT sizes of up to 524288 in this release.

The optimal size of the FFT for audio processing and graphic signal display are quite different. For this reason I kept the audio FFT size unchanged in this software, and added FFT size selection only to the graphical display code. The architecture of the pre-existing software made using the included SharpDSP FFT routines for this purpose difficult, and so I first included a hand-coded FFT routine. This code is contained in `fourier.cs`, which is included in source code I distributed, but is not used. I ended up using [FFTW](#). This is written for C, but there is a nice wrapper for C# use named [FFTWSharp](#). This did not include support for using Wisdom to optimize the FFT calculations, so I made a simple extension to the dll to permit Wisdom use. This extension merely consisted of adding the following lines to the code:

```
/// <summary>
/// W3SZ imports a wisdom plan
/// </summary>
/// <param name="plan">The plan to output</param>
[DllImport("libfftw3f-3.dll",
EntryPoint = "fftwf_import_wisdom_from_filename",
ExactSpelling = true,
CallingConvention = CallingConvention.Cdecl)]
public static extern int import_wisdom(string filename);

/// <summary>
/// W3SZ exports a wisdom plan
/// </summary>
/// <param name="plan">The plan to output</param>
[DllImport("libfftw3f-3.dll",
EntryPoint = "fftwf_export_wisdom_to_filename",
ExactSpelling = true,
CallingConvention = CallingConvention.Cdecl)]
public static extern int export_wisdom(string filename);
}
```

The modified DLL is included with my source code distribution, and the FFTWSharp source as modified by me is available [here](#). If the modified DLL is included in the project bin directory and appropriately referenced by the C# project, no additional coding is necessary to use the extended DLL.

2. Audio. As noted above, unlike PowerSDR, the original KISS Konsole did not provide a computer audio stream so that the user could listen to the HPSDR hardware using the computer audio system/speakers/headphones. Instead, one needed to obtain audio from the headphone jack of the HPSDR hardware [Mercury/Hermes/etc.]. I added computer audio output, first using the [Naudio](#) C# wrapper for Windows audio functions. Naudio can be added to C# using the [Nuget](#) Package Manager, which integrates very nicely into Visual Studio. Ultimately I decided that the [CSCore](#) audio package provided some advantages over Naudio, and so the final code uses CSCore audio rather than Naudio. The code to add audio to KISS Konsole was minimal, but it took me a bit of trial and error to optimize it. Receive audio is available at either the server computer, the client computer, or both depending on user selection. As noted above, the Opus codec is used to reduce both receive audio and mic audio ethernet bandwidth requirements. A server-side buffer for transmit audio has user-selectable size, ranging from 1 msec to 1000 msec, in case ethernet link conditions require a change from the default

value of 100 msec. The selection for this is at the bottom center of the Server Configuration setup page, shown above.

3. CW Sidetone. I also used CSCore to provide a CW sidetone. When doing remote operations, having a sidetone generated by the HPSDR hardware is unsatisfactory. So under those circumstances, the sidetone generated by the WinKeyer is used (or, alternatively, the sidetone generated by the client software at the operator's location is used).

4. Remote Capability. As dexteribed above, this software allows operation of the system from a remote location, using either wired or wireless ethernet communications. A computer microphone is used for transmit audio, a CW Key with or without WinKeyer is attached to a COM port of the computer running the client software at the operator's location and is used to generate CW, and a footswitch is attached to the same COM port and is used to provide convient MOX control, although the onscreen MOX button can also be used. Full operation of the station from 50 MHz through 24 GHz with automatic bandswitching is possible from a remote location. I use this software on a nearly daily basis to operate my remote station from home. Having the ability to use my station remotely has greatly increased my operating time.

**Summary.** "Always-On" bandscofes for each of the 50 MHz and up bands are essential to maintaining adequate situational awareness of band conditions and activity, particularly during contest activity. The openHPSDR hardware/software system provides a superb opportunity for the interested Amateur Radio Operator to delve into the world of Software-Defined Radios, providing opportunities for experimentation with both hardware and software.

I have presented a project for station automation and "Always-On" bandscofes for a station covering 50 MHz through 24 GHz based on the openHPSDR hardware and the KISS Konsole software. This provides for both local and remote operation, and eliminates the need for hardware switching of receive and transmit audio, CW keying input, and MOX (PTT) input.

The software presented is highly adapted for my particular station, and the intent is not for others to use the software presented as is, but to take the ideas presented here and create their own "Ideal Systems" for use at their stations, based on their particular circumstances. The locations from which the source code for the server and client can be downloaded are given below.

Roger Rehr  
W3SZ  
February 12, 2016

**References** are included as hyperlinks throughout the document.

There is further information on C Sharp SDR Controller and the Hardware SDR Controller on my web page <http://www.nitehawk.com/w3sz/CSharpsdrclientANDserverVersion2pt0.html> and you can download the source code for this software from links on that page, or from <http://www.nitehawk.com/w3sz/w3szclient.zip> for the client and from <http://www.nitehawk.com/w3sz/w3szserver.zip> for the server.