

# Elecraft K3 Operational Notes

## Serial: 2462

Compiled by Tony McClenny, N3ME

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**Modifications & Updates:**

Extreme Signal Protection			
IF Output Buffer			
KSYN3 Filter Terminator			
Added C101 on front panel			
Replaced AF/RF gain knobs			
Current Software			

## Transceiver Setup:

Heil Kenwood Interface Adapter (Red - HSTA-KM)

Connects to the lower left front panel "MIC" connector

Heil Proset Plus dual element headset

Microphone (Gray) mono plug goes into Heil Interface Adapter miniature mono jack

Headphone (Black) stereo plug goes into K3 rear panel "Phones" jack

## Transceiver Rear Panel Connections:

Antenna 1            Coax cable from TEN-TEC Titan Amplifier "RF Input"

Antenna 2            Currently not in use

AUX RF                Currently not in use

12 VDC In            Power cable from power supply

12 VDC Out           Currently not in use

Ground               # 6 Copper wire to radio room ground

RS232 Port            RS232 (Serial) cable to computer serial port 1

ACC Port              Currently not in use

SPKRS                Currently not in use

Phones                Headphone (Black) stereo plug

MIC Mono             Currently not in use

Line-In (Mono)	Line in to radio to computer sound card line out
Line - Out (Stereo)	Line out from radio to computer sound card line in
RX Ant (In)	Currently not in use
RX Ant (Out)	Currently not in use
XVTR In	Currently not in use
XVTR Out	Currently not in use
IF Out	Currently not in use
Paddle	Line to CW Paddle
Key	Currently not in use
PTT In	Line to Foot switch (using plug adapter)
Key Out	Line to TEN-TEC amplifier PTT/VOX controller
REF In	Capped - not used

## COM Port Sharing:

The K3 has a single RS-232 (COM) port. When you wish to use the computer to K3 COM port connection for general and/or contest logging software as well as CW, PSK, and RTTY, the single hardware COM port on the K3 limits your ability to have multiple COM ports talking to the K3, thus you need a hardware interface or a software solution.

Initially, I used a RIGblaster pro<sup>®</sup> hardware interface and it worked very nicely. However, as the K3 is designed to provide PSK and RTTY operation with the simple installation of two cables from the computer sound card to the K3 rear panel, it seemed like overkill to use such an interface and I sold the RIGblaster pro.

I use the N1MM Logger<sup>®</sup> program when contesting and the DX4Win<sup>®</sup> logging program for day-to-day contact logging. I discovered a software solution, which allows me to use the single K3 RS-232 (COM) port while operating multiple programs at the same time using "Virtual" COM ports.

N1MM Logger<sup>®</sup>                      Logging software

This software is available from:  
<http://www.n1mm.com>

DX4Win<sup>®</sup>                              Logging software

This software is available from:  
<http://www.dx4win.com>

Instructions for the use of PSK and RTTY are provided elsewhere in this document.

The next page tells how to obtain the software in use for "Virtual" COM port operation and how to configure it.

Another ham suggested I consider the use of "LP-Bridge" © software written and supported by Larry M. Phipps, N8LP. The software is provided as "Freeware" on his website located at: <http://www.telepostinc.com/LPB.html>.

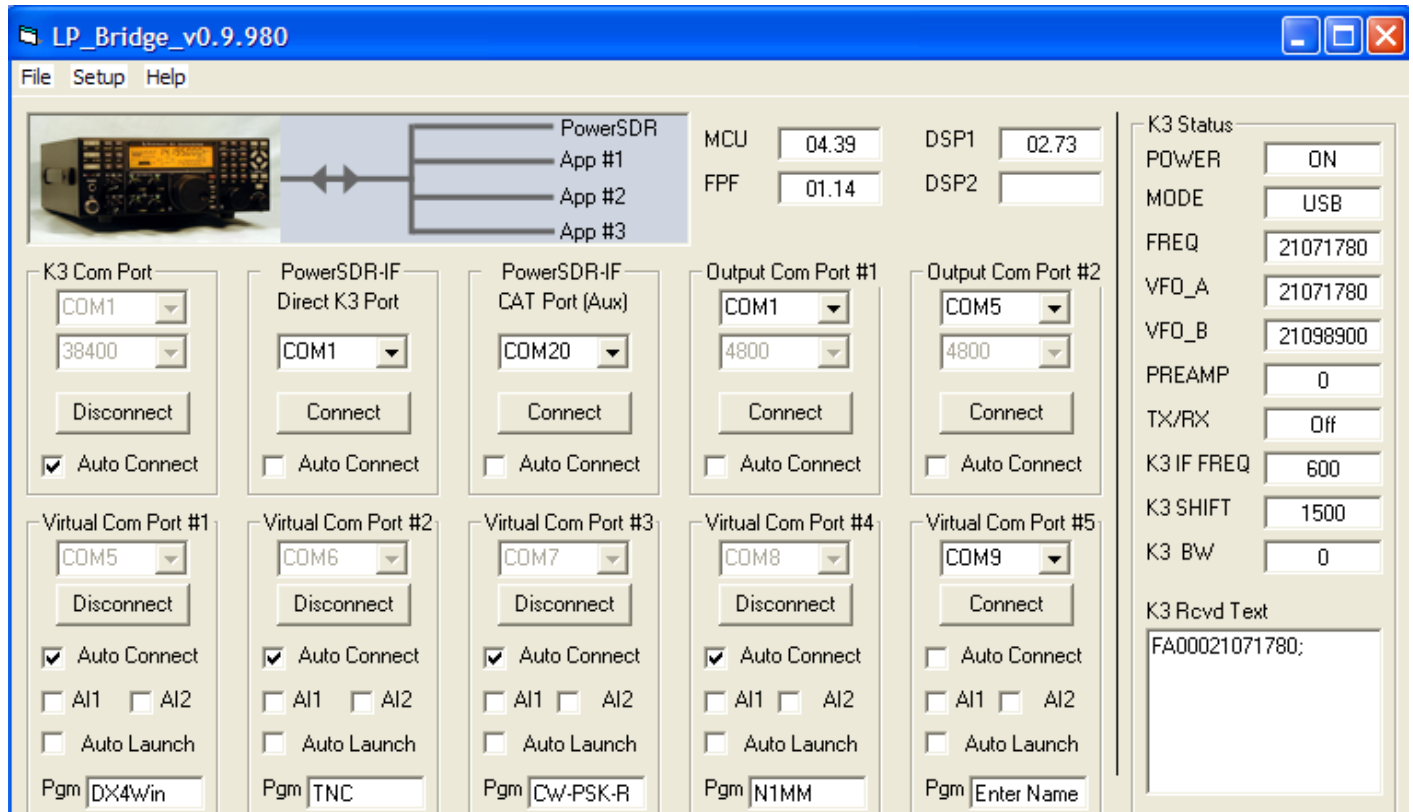
According to information posted on Larry's website, the LP-Bridge software is unique as it creates a mirror K3 in the computer's memory, with constantly updated data about the current state of the hardware."

Your first step should be to set your K3 to 38400 baud. This is accomplished by pressing and holding the "Menu" button on the front of the K3 until the "Config" window is displayed. Turn the VFO "B" knob clockwise until "RS232" is displayed and then turn the VFO "A" knob clockwise until "38400 b" is displayed. Finally, press the "Menu" button again to set these parameters.

Using Microsoft's Internet Explorer browser software or your personal choice of browser, go to <http://www.telepostinc.com/LPB.html> where you will find Larry provides a simple "click and download" capability of a "ZIP" type file so it is easy to obtain the software via the Internet

Once the software is downloaded to your computer, extract the files into a temporary folder and then run the software to install it on your computer. The LP-Bridge program will display a screen similar to the one shown below. Note: Default data has been modified in the picture shown in order to work on my computer.

This screen shot shows my current setup and I will comment on the various entries below the picture.



Note on the LP-Bridge screen display, you must set the "K3 Com Port" to an existing physical hardware COM port of your choice. Set the LP-Bridge software to the computer COM Port that is directly connected by RS-232 cable to the K3. The "K3 Com Port" must also be set to the same baud rate as the K3. The default setting on my computer happens to be COM1 but you may select any hardware COM port available in your computer by using the down arrow located on the right side of the "K3 Com Port" box located directly below the picture of the K3. You will notice 38400 is automatically displayed directly below the hardware COM port box.

Your next step is to use your mouse and left click on the "Connect" box, which is also located directly below the picture of the K3. If your radio is set properly to 38400 baud and you have selected an actual hardware COM1, COM2, COM3, etc. the "Connect" window will change to "Disconnect" and data showing the status of your K3 will be displayed in the far right column of the LP-Bridge window.

It is now time to setup up one or more "Virtual" Com Ports using the LP-Bridge software. During my installation, I setup "Virtual Com Port #1" to be COM 5 from which I operate my logging program (DX4Win), "Virtual Com Port #2" to be COM6 from which I operate various my Kantronics PacketCommunicator 3 Plus (TNC), "Virtual Com Port #3" to be COM7 from which I operate CW, PSK & RTTY, "Virtual Com Port #4" to be COM8 from which I operate N1MM.

Some of the software I use for various functions are shown below:

CW Decoder XP©                      Send and receive computer assisted CW

This software is available from:  
<http://www.ac6v.com/morseprograms.htm>

MMTTY©                                Send and receive computer assisted RTTY

This software is available from:  
<http://mmhamsoft.amateur-radio.ca/mmtty/>

N1MM©                                Contest log program

This software is available from:  
<http://www.n1mm.com>

WinPSK©                              Send and receive computer assisted PSK

This software is available from:  
<http://www.qsl.net/w2va/freeware.htm>

Instructions for the setup of each of these programs are available on my website:  
[www.n3me.net/howto.htm](http://www.n3me.net/howto.htm).

Each program must be installed on your computer and during the setup of each, select the "Virtual Com Port" port of your choice. You may select any COM port number as long as it is not an actual hardware COM port you may have installed in your computer. Remember, you only need one real hardware COM port in order to use this software. All the "Virtual Com Port" selections are not real computer hardware COM ports.

The LP-Bridge software will remember your setup of virtual ports, when used the next time. There are a few steps to take, when using the "LP-Bridge" program.

1. Boot up your computer
2. Turn On your K3 radio
3. Run LP-Bridge
4. Once all is running, you can use whichever ports you have setup.
5. For ease of use, I placed LP\_Bridge in my "Startup" file and it loads each time the computer boots up.

## My K3 Radio Setup:

The following connections have been made to my K3 radio to accommodate various external devices:

I am using a Heil Pro-Set Plus via a connection to the K3 Front Panel using the optional Heil AD-1 (Kenwood - Red) adapter cable.

The 1/8" mono plug from the Heil headset connects into the Heil AD-1 adapter, which is plugged into the K3 front panel "MIC" connector.

The 1/8" stereo plug from the Heil headset connects to the K3 rear panel "Phones" connector.

The foot switch that I use to control "PTT" on the K3 is connected via an adapter connector connected to the K3 rear panel "PTT IN".

My TEN-TEC Titan amplifier "PTT/VOX" connection is connected to my K3 rear panel "Key Out" via a short cable with an RCA connector on each end.

My MFJ564 CW key is connected to the K3 rear panel "Paddle".

To properly configure the K3, briefly press and release the "MENU" button to access the "Main" menu selections.

Turn the "VFO B" knob to:

Set "MIC SEL" FP.L

Briefly tap the "A/B" "1" "B SET" button to select .Low or .High microphone gain range.

Briefly tap the "REV" "2" button to select "BIAS" "off" for the Heil Pro-Set Plus headset.

Set "VOX GN" 50

Press and hold the "MENU" button to access the "Config" menu selections.

Set "AF Gain" High

As I wish to hear through both the Heil headset and the internal K3 speaker at the same time, I made the following "Config" selections:

Set "SPKRS" 2

Set "SPKR+PH" Yes

Briefly press the "MENU" button and the settings will be stored in memory.

Press the "MODE" button until USB is displayed on the K3. Briefly press the "MENU" button and "MAIN" will display on the K3 display. Turn the VFO B knob until "MIC SEL" is displayed on the K3. Turn the VFO A knob until "FP.L" is displayed.

Briefly press the "MENU" button and the setting will be stored in memory.

The K3 rear panel RS232 port is connected to my computer COM Port 1 via a 9 pin serial cable.

If you use the USB / Serial adapter from Elecraft, it is probably an FTDI or Prolific device. Disconnect your USB cable from the computer, install the correct driver and after the driver installs, then plug in the cable. After you plug in the cable, go back into Control Panel and see

what com port # Windows has assigned to your adapter. Set your software to match that com port number.

If it is an FTDI, go here and download the correct driver:

<http://www.ftdichip.com/Drivers/VCP.htm>

If it is Prolific, go here and download the correct driver:

<http://www.prolific.com.tw/eng/downloads.asp?ID=31>

### **Power "Gain" Calibration:**

The following comment comes from an e-mail dated February 26, 2009 written by Wayne, N6KR.

There isn't any "100 W calibration" on the K3. All gain calibration is done at 5 and 50 W (and optionally 1 mW if you have a KXV3). To calibrate the rig is at any setting from 13 to 100 W depends on whether you've done the 50 W calibration.

There is one additional form of optional calibration: the K3's wattmeter.

See CONFIG:WMTR in the owner's manual.

There is a procedure on page 49 on the K3 manual to calibrate the power out on the K3. After doing this you calibrate "per-band" using a procedure on the same page. The K3 power meter is not absolute. In other words, this procedure on page 49 is what tells the K3 power meter how much 100 watts really is (actually 50 watts is used for the test).

## PSK and RTTY operation:

1. Turn the Elecraft K3 transceiver "Off".
2. Connect "Line In" port on rear of transceiver to computer "Line Out" using a monaural cable.
3. Connect "Line Out" port on rear of transceiver to computer "Line In" using a stereo cable.
4. Turn the Elecraft K3 transceiver "On".
5. You can use either PTT or VOX by setting the K3 "CONFIG PTT—KEY" to either RTS-off or DTR-off. If using PTT, the K3 will be placed in transmit mode when the computer is started as it will activate the signal in the serial port.
6. Press the "MENU" button briefly and "MAIN" will display on K3  
Turn the "VFO B" knob until "MIC SEL" is displayed  
Turn the "VFO A" knob until "Line In" is displayed  
Press the "MENU" button briefly to turn off the "MENU"
7. Press and hold the "MENU" button until "CONFIG" is displayed on the K3  
Turn the "VFO B" knob until "RS232" is displayed  
Turn the "VFO A" knob until "38400 b" is displayed  
Press the "MENU" button briefly to turn off the "MENU"
8. Turn the "CMP PWR" knob on the K3 transceiver to  $\frac{1}{2}$  maximum transmit power level  
Press the "VOX" button to set "VOX" to "On"  
Press the "MODE" button repeatedly to set "MODE" to "DATA"  
Press and hold the "DATA MD" button until "45 bPS" is displayed on the K3  
Turn the "VFO B" knob until "AFSK A" is displayed on the K3  
Briefly press and release the "DATA MD" button  
The transceiver is now set for "AFSK RTTY" operation for as long as the "MODE" remains set to "TX DATA".

9. Press and hold the "PITCH" button and set "MARK/SHIFT" by turning the "VFO A" knob. It is suggested to use "2125 - 170".
10. For PSK31 (Set Mode to DATA A) or RTTY software so you can test your K3 settings. DATA A uses the upper sideband and the dial displays the suppressed carrier frequency.
11. Press the "TEST" button so the K3 display shows, "TX TEST". Then transmit PSK or RTTY and view the ALC Meter. The level of input from the computer sound card should display 4 solid bars and the 5<sup>th</sup> bar should blink on and off. The fifth bar indicates the onset of ALC. If your level is below or above this display level, turn the "CMP/PWR" knob to adjust to this level. At present, the level is "6".

When the level is set correctly, press the "TEST" button to return the K3 to "TX NORM".

Note: You may need to increase the computer sound card output volume. Do this in stages until you find the correct setting for your computer/radio combination.

Note -- if you have more than one sound card, the normal volume settings displayed apply only to the default soundcard. You must use the Control Panel to temporarily change the default soundcard to adjust the controls on the second sound card, and then change the default back.

12. Select the desired filter(s) using the "XFIL" button.
13. Once you have accomplished steps 1 - 8, you are ready to load the digital software of your choice for CW, PSK or RTTY operation. Here is a short list of programs you can use with your newly created virtual ports:

CW Decoder XP©                      Send and receive computer assisted CW

This software is available from:  
<http://www.ac6v.com/morseprograms.htm>

MMTTY©                                Send and receive computer assisted RTTY

This software is available from:  
<http://mmhamsoft.amateur-radio.ca/mmtty/>

WinPSK©

Send and receive computer assisted PSK

This software is available from:

<http://www.qsl.net/w2va/freeware.htm>

Instructions for the setup of each of these programs are available on my website:

[www.n3me.net](http://www.n3me.net).

### K3 Breakout Box:

Bud Governale provided a wiring diagram for a breakout box that he made to capture all the possible needed connections that come from the K3 accessory connector. Using this box, one can use two COM ports with RS232 cables; one for radio control and one for FSK RTTY keying. The diagram is available as a .pdf file from my website: [www.n3me.net/pdf/K3 Breakout Box wiring diagram](http://www.n3me.net/pdf/K3%20Breakout%20Box%20wiring%20diagram.pdf).

I elected to use the LP\_Bridge software, which creates virtual com ports.

## PSK, AFSK and FSK Mode Programming Information:

To use the DATA A mode, first turn VOX to "On" as it is set "Off" by default.

These comments are taken from an e-mail written on April 2, 2009 by Joe Subich, W4TV.

"That depends on how you define "PSK" mode ... the Elecraft K3 support audio based data modes useful for PSK.

The Elecraft K3 has separate PSK, AFSK and FSK modes.

The K3 needs to be operated in extended CAT mode (K31 On). The CAT "data" modes are "normal" (MD6;) and Reverse (MD9;)

The data submode is set with the DT; where:

DATA A = DT0

AFSK A = DT1

FSK D = DT2

PSK D = DT3

; Note: PSK D is only useful with the KY; command.

For FSK D and AFSK A, "normal" is LSB and "reverse" is USB.

For Data A and PSK D, "normal" is USB and "reverse" is LSB.

The data submode is indicated in the 35th byte of the IF response."

When using PSK31 and similar modes, the K3 VFO A is set to a frequency for the carrier. The audio is on the upper sideband, and is centered in the 3 KHz passband which is 1.5 kHz up from the carrier - thus the offset.

SSB Audio - To properly adjust the audio into the K3, check that the ALC meter is lighting up 5 to 7 bars with normal speech. If not, set the compression to zero (1:1) and use "TX TEST" so you don't transmit. Adjust the "SPEED MIC" (microphone gain) knob so the ALC meter indicates at least 5 bars and peaks at 7 bars. When this process is complete, increase the compression as desired after which turn off "TX TEST".

To properly adjust the power output for CW or Digital use the Mode button to select the mode you wish to set and the "SPEED/MIC" button to view and adjust the MIC or Line Out power/drive. The settings will display on the "RF" bar graph in the display window.

To make this Line Out adjustment for digital modes, place the K3 in TX Data Mode, press the "SPEED/MIC" knob twice and adjust the power output while watching the display. Adjust the audio input until you see 4 solid bars and an occasional flicker of the 5th bar. My "SPEED/MIC" control is set to 8 for digital modes.

To make the MIC adjustment for CW or SSB, place the K3 in CW, LSB or USB Mode, press the "SPEED/MIC" knob twice and adjust the power output while watching the display. My "SPEED/MIC" control is set to 8 for digital modes and 20 for SSB/CW.

For voice modes, first turn off compression, then go up until the 7<sup>th</sup> bar flickers. After setting the audio level, then add back compression as desired. These steps should be done in "TX TEST" so no RF is transmitted.

The display is not an indication of ALC action or compression. The ALC does not start until the 5th or 6th bar is illuminated - the lower 4 bars are an indicator of the audio level (sort of like a Vu meter).

## Headset & Microphone Information:

I am using the Heil Pro Headset microphone connection via the Front Panel connection using the optional Heil AD-1 Kenwood (Red) adapter cable.

The 1/8" mono plug from the Heil headset connects into the Heil AD-1 adapter, which is plugged into the K3 front panel "MIC" connector.

The 1/8" stereo plug from the Heil headset connects to the K3 rear panel "Phones" connector.

To properly configure the K3 for this setup, briefly press and release the "MENU" button to access the "Main" menu selections.

Turn the "VFO B" knob to:

Set "MIC SEL"            FP.L

Set "VOX GN"            50

Press and hold the "MENU" button to access the "Config" menu selections.

Set "AF Gain"            High

As I wish to hear through both the Heil headset and the internal K3 speaker at the same time, I made the following "Config" selections:

Set "SPKRS"            2

Set "SPKR+PH"          Yes

As I am using a standard Heil headset (with dynamic microphone elements HC4 & HC5), I have "bias" turned off.

The "H" or "L" refers to the microphone gain range and \*not\* to the bias level. You may want to try the low range to have more control over your microphone gain (your microphone level setting will require higher numbers).

To temporarily increase the headphone volume, open the K3 Utility program and choose "Command Tester". Enter the following code on the top line and hit "Enter": !66; You can do this several times and the volume will increase slightly each time. This setting is NOT retained, if you shutdown and restart the K3.

Settings depend on one's voice and how you want to sound. Here are three combinations I've tried:

	Prelim	Final	Heil
Mic Gain Range	FP.h	FP.L	FP.L
Mic Gain	30	35	25
Comp	20	23	23
VOX Gain	050	050	
VOX Delay	15	15	

0.05	-16	-12	-12
0.1	-16	-10	-10
0.2	-4	-6	-6
0.4	0	-2	-2
0.8	0	0	0
1.6	+6	+6	+6
2.4	+8	+8	+8
3.2	+8	+10	+10

The rear panel microphone jack is a stereo (T-R-S) jack with the mono audio mic signal connection on the tip and the shield should be connected to the shell. The ring is not connected. It is an unbalanced input and accepts balanced or unbalanced dynamic mics but only unbalanced electret condenser mics. When RL.X bias is on, it supplies 8 volts.

Heil published the following information in their website article, "DSP Settings and Mic Connections: Elecraft K3.

### DSP Settings and Mic Connections: Elecraft K3

Note: These are starting points only. The best setup for your voice, your microphone, and your operating objectives may differ. Be sure to listen to yourself on a separate receiver (with its antenna disconnected and noise blanker turned off) as you make final adjustments to optimize the settings for your operating situation.

#### Connections:

The K3 utilizes the Kenwood protocol for connections to its front panel 8-pin microphone jack, so use the CC-1-K adapter for 4-pin XLR microphones (GM Series, HM-10 Dual, and Handi Mic), and use the CC-1-XLR-K adapter for connection to 3-pin XLR microphones (PR-781, etc.).

Briefly press the "Menu" button on the K3 front panel so you will be placed in the "Main" menu selections. Set the "MIC SEL" menu selection to "FP.L" (Front Panel).

For Heil headsets, use the Heil AD-1-K for connection to the front panel "MIC" jack.

Briefly press the "Menu" button on the K3 front panel so you will be placed in the "Main" menu selections. Set the "MIC SEL" menu selection to "FP.L" (Front Panel).

The K3 also has provision for direct connection of the microphone cable from a Heil Sound headset, with no adapter required, if you use the K3's rear panel "MIC MONO". If this is the connection you prefer, briefly press the "Menu" button on the K3 front panel so you will be placed in the "Main" menu selections. Set the "MIC SEL" menu selection to RP (Rear Panel).

See pages 13, 17 and 20 of the K3 Operating Manual for more details.

## What Kind of Mic?

Most operators will be connecting a dynamic element microphone (HC-4, HC-5, or PR-781 type). In this case, while setting up the MAIN: MIC SEL menu, tap [1] to toggle to "LOW" Mic Gain, and be absolutely certain that Mic Bias is off by tapping [2], as needed, to toggle to "OFF" for Mic Bias.

If you are connecting to an "iC" element (Pro Set iC, Pro set Plus iC, Pro Set Elite iC, Handi Mic iC, or the iCM via the HSTA-K8 adapter cable), tap [1] to select "HIGH" Mic Gain, and tap [2] to set the Mic Bias to the "ON" option. See pages 13 and 20 of the K3 Operating Manual.

## Eight-Band Equalizer Settings

The K3 includes a versatile 8-band transmit equalizer, which is adjusted separately from the receiver equalizer. The K3 also has an "ESSB" mode, with a wider transmitted bandwidth for high fidelity operating. Please read pages 35-36, and page 60, of the K3 Operating Manual, in particular, for more details of these subjects. *Yes, you have to read the manual!*

The table that follows details *starting points* for operation using two main categories of mics: the "articulated" elements (HC-4 and HC-5 based mics), and the "wide range" elements, typified by the iCM, Handi Mic iC, and PR 781 and the PR 20/30/40 genre. Remember to set the Mic Gain and Mic Bias properly, as described above.

## Eight Band Equalizer Starting Points

Freq >	0.05	0.1	0.2	0.4	0.8	1.6	2.4	3.2
HC-4/5	-10	-4	-2	0	0	+2	+4	+4
Wide	-12	-10	-6	-2	0	+6	+8	+10

The above settings are for "normal" operation. For ESSB operation, you may want to set the lower-frequency selections to *positive* settings, depending on which microphone you have and your audio sound objectives.

Again, these are necessarily *starting points* for operation, and it should be noted that, when using the HC-4 and HC-5 elements, the equalizer may be left at its default setting of "0" across the board, since equalization is already taking place in the mic element itself.

## Mic Gain and VOX Setup

Follow the instructions on pages 28 and 29 of the K3 Operating Manual. There are many inter-dependent adjustments on the K3, and the setup instructions on those pages are excellent.

The following was provided by Olli (DH8BQA):

No problem at all and no dream anymore, the [www.remoterig.com](http://www.remoterig.com) solution has everything you mention already built-in, from VoIP to an own CW keyer also taking care of the internet delays. Works like a charm (am using it with a IC-706 with the control head being 700 km away from the rest of the radio and everything else) and no problems even with 80 ms latency (read round trip time). Full control over the radio, exactly the same functionality as if the control head was directly connected to the radio. "I" have made almost 1.500 QSOs with that solution during the last 2 years (not that active due to QRL reasons).

Besides "modes" in their controllers for remoting the IC-706's or TS-480's heads over the internet they also have profiles for using it with serial connections for radio control, i.e. CAT. So a "self-made radio head" is no problem either, have tried that "half-way" myself already with a small PIC display I built for my FT-817. So just create a front panel, send all button presses as CAT commands to the remote radio, connect the microphone, key and speaker directly to the remote rig controller and et voila, ready's your remote station with an Elecraft remote head.

You can do that with a CAT software on the PC, too, i.e. Ham Radio Deluxe, but then you'll loose the "feeling of spinning the dial" that's why I opted for the remote head solution with the IC-706 although the radio is really not the best (and no comparison to my K3, of course).

I think this would be a real seller! Kenwood recently started producing and selling remote heads for their TS-480's seperately besides "complete" transceivers as people are starting to share their remote solutions so you need 3 or 4 heads for 3 or 4 guys to access the same radio (not at once, of course).

## ATU Operation:

The KAT3 ATU mode is normally set to *BYP* (Bypass) or *AUTO* (Automatic) outside the menu. ATU alternates between the two modes.

Modes *L1-L8*, *C1-C8*, and *Ct* are used to test KAT3 relays.

Mode *LCSET* allows manual adjustment of *L/C/net* settings (you must exit the menu first). When in *LCSET* mode, tapping *ATU TUNE* shows the *L* and *C* values; *C* is changed with *VFO A*, *L* is changed with *VFO B*, and *ANT* toggles between *Ca* and *Ct* .

Tapping *CLR* within this menu entry clears stored *LC* data for the present band.

The following provided by Fred Jensen (K6DGW):

""Automatic" is a word with lots of meanings. When it is in "AUTO" and you tap *ATU TUNE*, it will "automatically" find a match using the power you have set in *CONFIG:TUN PWR*. Mine is set to 5W. It then "remembers" the *L/C* settings for that band segment for the *ANT* selection you are using [1 or 2]. The next time you are in that band segment using that *ANT* selection, it recalls the settings the first time you transmit [*PTT* or *KEY*].

Push the *ATU TUNE* button once to get a quick but somewhat rough *SWR* match. If you push *ATU TUNE* a second within 5 seconds of the first pass completing, the ATU will take more time and come up with a better *SWR* match.

The band segment sizes depend on the band. I believe they are:

160: 10 KHz  
80-12: 20 KHz  
10: 100 KHz  
6: 200 KHz

You can just let the memorized settings accrue as you operate normally, or you can methodically go through the bands and set them all. I did the latter, tuning to the middle of each segment with the antenna I normally use, and tapping *ATU TUNE*. Seems to work really good."

It is also engaged/bypassed per band per antenna according to the remembered setting on that band/antenna combination.

## AGC Discussion:

AGC means automatic gain control and its function is to keep the receiver's audio output more or less constant as the RF input signal changes level.

It is suggested to use slow AGC during a contest perhaps set to its fastest setting - this will keep the loud signals linearly loud and short enough to recover when the loud station is worked and moves off frequency.

Dave Hachadorian (K6LL) suggests,

1. Dedicate an F-Key in your logging program to turn off agc whenever a loud pileup shows up. Turn it off on both rigs if you are so2r. In N1MM, the macro is:

```
{CAT1ASC SWH027;}{CAT2ASC SWH027;}
```

2. Add to the "exchange" message a command to turn fast agc back on (for both rigs in so2r). In N1MM, the macro is:

```
{CAT1ASC GT002;}{CAT2ASC GT002;}
```

The effect of the above is that agc is turned off, and gain is reduced, only while you are struggling to pick a callsign out of a loud pileup. As soon as you send the exchange, fast agc is restored, and gain goes back up on both rigs.

The following is a quote from an e-mail written by Wayne, N6KR and Lyle, KK7P on February 25, 2009.

Details:

Some operators described a "merging" of multiple, close-spaced signals when listening to pileups, making individual signals hard to copy. Assuming that a narrow crystal filter is in use (close to the DSP bandwidth), this is not "desense". In most cases, we've discovered, the cause is IMD (intermodulation distortion) related to the AGC algorithm.

To simulate what happens under pileup conditions, we combined four crystal oscillators with gaps between them of 5, 6, and 7 Hz (a total spread of 18 Hz for the four signals). We then injected the signals into the K3 at an equal level of about S5 and used spectral analysis to

examine what happens to the passband. With AGC-F turned on (AGC fast, set to the K3's factory defaults), IMD products came up some 20-25 dB as compared to AGC OFF. The situation could be far worse in an actual on-air situation with more signals, noise, and key clicks. (AGC-S makes only a small improvement.) We ran the same test on several other radios, and verified the AGC-induced IMD characteristics are similar.

IMD is generated because AGC can cause mixing between the tones and their harmonics. The faster the decay rate, in general, the more products will be generated. With just one or two signals, most products will be outside the DSP's narrow IF or AF passband. But in a pileup, where signals are greater in number and can be nearly on top of each other, there are many more opportunities for signals to combine. This results in IMD products spaced at 1 to 20 Hz, depending on the time constants, location of the signals, etc.

Turning AGC off is one strategy that operators have traditionally used to combat the effects of AGC modulation. It then becomes necessary to "ride" the gain controls. Neither Lyle nor I have that much patience :) so we took a different approach: keep the AGC on, but reduce its modulation effects to negligible levels by adding a very long "hold" time.

In the field-test firmware, two changes have been made. First, the AGC HLD menu setting, which used to apply only to voice modes, now applies to CW as well. (It still only affects AGC-S, the slow AGC setting.) Second, we increased the range of AGC HLD to facilitate experimentation. What we found is that an AGC hold time of about 1 second, combined with a slower AGC-S decay time, reduces in-band IMD with closely spaced signals by 20-25 dB, making it very similar to turning AGC off. The \*attack\* time is unaffected, so there is no "thumping" as you might have with audio-derived AGC. Any new signal peak above the one that starts the "hold" timer will restart the timer.

This "hold" AGC appears to be useful in many listening situations. It can make voice signals sound cleaner, as well; after all, a voice signal can have many components, and AGC-induced modulation can add IMD products. It also helps with noisy band conditions.

The down side to AGC "hold" is that large signals can charge it up to a high level, reducing gain for the duration of the hold time. But in many cases this is far preferable to the AGC IMD effects, especially if you're listening to many signals at similar amplitudes -- as in a pileup.

You'll want to play around with agc slope and threshold. I settled on a slope of 10 and threshold of 8. Discover what settings you prefer. Don't forget the receive equalizer either which is a very nice feature. I've found not all speakers or phones sound the same. It's nice to tweak them with the equalizer.

The following was written by Bill W4ZV:

Note that Eric's setup is for AGC Off, which is probably not a good idea for most casual users. I would start with AGC-F using the default AGC Menu settings. You can check the defaults for each CONFIG: AGC Menu by tapping the DISP button while in the appropriate CONFIG Menu (all AGC settings begin with CONFIG: AGC xxx, and remember that some are only accessible with CONFIG: TECH MD On).

1. Have available your desired antenna and a dummy load (both on a coax switch is ideal) for each band you want to set up (you'll at least need to do this for major segments like 80/160, 40-18, 15-10 and 6 separately (if you have the external preamp).
2. Set AF GAIN back to about 9 o'clock and RF GAIN to MAX (fully clockwise). This assumes you have the proper AF GAIN menu setting for your specific headphones ( I use AF GAIN - LO for my 32 ohm headphones and still have tons of audio).
3. In the following order (first ATT ON, then ATT OFF, then PRE ON) alternately switch between your antenna and the dummy load. When you hear noise increase when switching to the antenna, use the lowest gain setting (i.e. ATT will likely be the correct setting on the noisy low bands and PRE will likely be correct for 10m). Remember that the K3 will remember these settings for each band. Once you have the correct setup on each band it will automatically recall that PRE/ATT setting when you return to that band.
4. Now adjust the RF GAIN knob counterclockwise until the antenna noise on the meter just stops flickering on the meter. This will be familiar to OTs since this is the way we set RF GAIN many years ago on analog radios.
5. Now adjust your AF GAIN knob for a comfortable listening level (mine is normally 9-10 o'clock which may vary depending on what type of headphones you have).

If you still don't like the amount of background noise compared to signal levels, adjust CONFIG: AGC THR to a lower setting than the default 5. This which will reduce the apparent noise level by moving the AGC onset point down and acting as sort of a noise squelch.

Regarding settings for speakers, I never use them since I'm normally listening for extremely weak signals on 160m or in contests and I don't want ANY extraneous noise between my ears and the signals. Perhaps someone who uses speakers can add comments.

I hope the above helps but please read K3NA's article for more comprehensive suggestions (remembering that his setup is for AGC Off).

My general rule for all gain settings (PRE/OFF/ATT, RF Gain knob, AF Gain knob, CONFIG: AF GAIN) is LESS IS BETTER (as long as you can hear signals). This is contrary to the thinking of some but following it will ensure that you get the ultimate dynamic range performance available from the K3.

Further comments from Wayne (N6KR) on AGC:

The K3 provides two alternatives to turning off AGC in this situation.

We closely examined the effect of these on multiple close-spaced signals, and both reduce IMD considerably:

1. CONFIG:AGC DCY sets the AGC decay characteristic, and applies to both AGC-S and AGC-F (slow and fast AGC). The default setting NOR, provides a traditional linear decay. The SOFT setting applies a "soft" exponential decay. It typically reduces in-band IMD by 10-15 dB.

1. CONFIG:AGC HLD adds a "hold" timer to slow AGC-S (slow AGC) only.

It behaves as if you were manually (and very quickly) riding the RF GAIN control, reducing gain when a very strong signal comes in, then turning it back up N seconds later. The normal setting is 0.00 (no hold time); a typical setting for pileup conditions might be 0.40.

This can reduce in-band IMD as much as turning AGC off.

Andy (AE6Y/P49Y) makes the following comments:

"I would add that I have found that using AGC-S with SLP=6 and THR=6 gives me the ability in CW pileups to separate out the signals by riding the RF gain control, without turning off the AVC and risking getting blown out by strong signals. I used these settings in WPX CW this past weekend and found them very effective. It's hard to figure this out on the fly, but I had tried varying the settings by listening to another station's pileup a few contests ago.

The K3 is just a great run radio: unless the adjacent stations have clicks, you simply don't hear them (though the P3 shows just where they are, if you want to optimize your frequency between big signals)."

## AGC Settings:

To modify the factory default *AGC* settings, press and hold the "Menu" button until "Config" is shown on the K3 display. Turn the "VFO B" knob until "AGC DCY" is displayed. Then turn the "VFO A" knob to the setting you prefer.

Dave, K6LL, and Ignacy, NO9E made the following suggestions for *AGC* settings for general use in e-mail messages.

	Dave, K6LL	Ignacy, NO9E	N3ME
DCY	Soft	Nor	Soft
HLD	0.20	0.50	0.05
PLS	nor	nor	Off
SLP	0	010	0.07
THR	8	005	007
F	200	150	080
S	020	020	020



### Carrying Case:

A Mil-spec aluminum case (#138401) from Coleman's Military Surplus ([www.colemans.com](http://www.colemans.com)) is available to accommodate an Elecraft K3 transceiver for heavy-duty transportation. Thanks to Dave Olean, K1WHS, for finding this case and posting its availability to the Elecraft reflector on November 13, 2009.

## Mailing List Guidelines:

### Elecraft Mail LIST GUIDELINES

For those of you who are new to the list, (and for those of us who have rapidly failing memories..), here is a quick list of things to remember when posting to this list. Please save this for future reference.

The most important thing to remember is that this is only a hobby - Let's have fun!

1. YOU MUST BE SUBSCRIBED to the [Elecraft] list TO POST to it. (This is done to stop advertising spammers from hitting the list.) Any postings sent to [elecraft@mailman.qth.net](mailto:elecraft@mailman.qth.net) by addresses different from the exact ones it shows as subscribers will be rejected.

This includes alias (forwarded) addresses like [w1xyz@arrl.net](mailto:w1xyz@arrl.net). If you use an alias to subscribe you must have it as your from: and return address too.

Subscribing with [w1aw@arrl.net](mailto:w1aw@arrl.net) from your physical address of [joe@aol.com](mailto:joe@aol.com) will allow you to receive postings, but your postings to the list will be rejected if their from: and reply to: address does not match your subscribe address..

Go to <http://mailman.qth.net/mailman/listinfo/elecraft> to subscribe and to change your list preferences. To unsubscribe or to change your list preferences (digest, no mail on/off etc.), scroll to the bottom of the page and log in with your subscribed email address and the password that was sent to you by email when you subscribed (and monthly afterwards.)

2. If you want to provide an attachment, .JPG picture or other large file for use on the list, first post it to your personal web page and then post a link to its address in an email to the list. The list strips all attachments to prevent viruses from propagating and to keep the archives at a reasonable size.

3. Please keep your postings short and the amount of copied text from previous posts to an - ABSOLUTE MINIMUM- in your replies.

Always delete everything from the prior post except what is necessary to keep your reply in context.

Most copied messages can be reduced to one or two sentences to retain context.

Remember to delete the email list footer from the previous post.

Especially try to avoid copying a long prior posting and adding 'Me Too!' or something similar. As the number of users on this list grows (over 3500 now) we need to work to minimize information overload... If a reply is not of interest to the list, just reply directly to the posting party.

#### 4. EMAIL OVERLOAD:

If you are overloaded by the volume of individual messages on the list, you can view the searchable daily Elecraft list messages for each month in web format at:

<http://www.elecraft.com/elist.html> . These archives are updated hourly and list all postings by subject. Just click on the ones you are interested into read.

You can also set your list email preferences to 'no mail' delivery, which still allows you to post to the list when reading via the digest.

You can also change your subscription to the DIGEST version, which sends you a single compilation each day.

To change your email list options or to subscribe / unsubscribe, go to:

<http://mailman.qth.net/mailman/listinfo/elecraft>

Scroll to the bottom of the page to log into your preferences page and set your mail options to 'no mail'.

4a. Please make an effort to keep list volume under control by resisting the urge to post a comment on every long discussion thread (CW, Band Switching, and Soldering etc.) With over 3,000 list subscribers volume can quickly get out of control if everyone feels the need to comment.

While we do not overly restrict the subject matter on this list, and we encourage postings on a wide range of amateur radio related topics, please remember that the list's primary focus is on Elecraft products and their use.

Many people rely on it for pointers on building, using and troubleshooting their Elecraft products.

4b. \*\*\* [NEW] When emailing about a specific rig or option, please add the rig/option name(s) to the first part of your email subject line. (K1, K2, K3, KX1 etc.) This will be a huge help for those experiencing email overload and will allow automatic filtering based on subject line.

Examples: "Subject: [K3] Filter Options"

"Subject: [KX1] How to use ped portable?"

"Subject: [XG2] Wow! It's a big help.

5. \*\*\* IMPORTANT \*\*\* PLEASE KEEP ALL POSTINGS CORDIAL.

Restrain the urge to email someone admonishing them about a posting.

The last thing we want to do is to scare anyone off the list. Overly aggressive postings and negative comments about other posters only serve to scare away new potential list members. Waiting over night before hitting 'send' really helps to put things in perspective. Please make all postings as if you were talking face to face to a close friend you do not wish to offend.

If you have a complaint about someone's posting, spam or a thread topic please email directly to me ( [eric\(at\)elecraft.com](mailto:eric(at)elecraft.com) ) and I'll address it.

Do not post your complaints to the list.

5a. Please -do not- post publicly or privately asking people to stop a particular thread, no matter how long, off topic or repetitive it gets. Email me instead ;-).

I will step in when I feel it is necessary to end a thread. ([eric\(at\)elecraft.com](mailto:eric(at)elecraft.com))

5a1. Please exercise restraint in posting when a thread is getting heavily covered. 10 to 20+ posts on one topic in a day are usually excessive. Please try to stop posting to any thread, especially OT threads, once 10+ replies have been posted.

5b. Do not post any direct attacks or snide comments directed at a list member. Enthusiastic arguments are encouraged, but please keep everything cordial. Members who verbally attack another (either via the list or via direct email) will be removed from the list.

As the 'official' list moderator, I'll jump in as necessary to keep everything orderly. I do this off line and occasionally to the list when it is appropriate. Our goal is to keep the

Elecraft list a fun, informative central clearing house for Elecraft information and enthusiasm.

6. Please DO POST your technical questions and comments to the list.

Elecraft owners are your best source for quick answers (and they NEVER sleep!) If you don't get the answers you need from the list please email us direct at support(at)elecraft.com , which is our customer service address. We do try to watch the list traffic, but we may not respond to everything immediately and may miss some postings sent to the list or our personal addresses.

7. Please post your experiences with your Elecraft kit, DX worked, crazy ideas, product ideas, complaints (yes - we do want to hear them). OT-Ham Radio Related posts are OK, but please keep them short and focused. (See above about limiting OT thread replies.)

7. Commercial postings are allowed if they relate to Elecraft products, QRP, QRO, home-brewing, building etc. and are of interest to this list's membership. Please keep them as short as possible and provide web links to more detailed information. I'll step in if we feel someone is posting too many messages of this type.

7a. Please limit commercial postings to one per month per offering or product area (i.e. Builder for Hire postings, ham radio related products etc.)

8. Once again, personal attacks, flames, or strongly worded derogatory messages will not be tolerated. (Pausing overnight before pressing the send key really helps..) Please do not criticize postings by others.

9. Send parts requests direct to parts(at)elecraft.com, not to the list. (Replace '(at)' with '@' in this address.)

10. If you don't get an answer to a question from the list, or by checking the list archives, don't forget to check the Builder's Resource Page at <http://www.elecraft.com> and our support email addresses: support(at)elecraft.com (non-K3 products) and k3support(at)elecraft.com (K3). (Replace '(at)' with '@' in these addresses.)

11. And above all, LET'S HAVE FUN!. We hope that this list is both a good technical resource and serves as an Elecraft community gathering place. We enjoy reading it every day and it

really helps us keep our energy level high so we can continue getting exciting new products out the door to you!

73, Eric WA6HHQ  
Elecraft List Moderator

### **Mailing List Password:**

"If you have misplaced your Elecraft list password, or if you never received it, here is how to get it sent to you again.

Go to: <http://mailman.gth.net/mailman/listinfo/elecraft>

and scroll to the bottom of the page. Enter your email address (the one you used for the list) and click on 'Edit Options'. This takes you to your personal list options page. One of the buttons on that page is labeled" 'Email my password to me'.

Once you have your password, go to the above link and enter it to change your subscription parameters (digest mode, unsub etc.)"

73, Eric WA6HHQ

Don Wilhelm (W3FPR) writes, "The K3 noise blanker... works by producing "holes" in the receive path. That is what blanking means - if an impulse is detected, it shuts off the receive path for a specified period of time.

That will not "take away" from the reception of faint signals, but if combined with gaps in the received signals because of impulse events occurring at the receiving station, it may make the copy of faint signals difficult because of the blanks in the received data stream.

In my opinion, it should only be used when necessary, not all the time.

If you have constant static pulses or automotive ignition noise or distant lightning surge problems at your QTH, then it may be beneficial to run the NB at all times, but if you do not need it, keep it off.

Remember that the NB responds to impulse noise - it will not be effective in reducing slow rise time noise like atmospheric noise or power-line induced noise. NB needs a fast rise time to be effective.

Noise reduction (NR) will reduce or eliminate things like atmospheric noise or power line noise as it works on correlation principles. This means it is normally quiet while it looks for what might be a valid signal. When it finds what appears to be a valid signal, it builds a filter around it and sends the result to the K3 audio. What it looks for is highly dependent on the parameters set for the K3 NR algorithm as well as your local noise. Experiment with your chosen mode and local noise situation until you discover what works best for you. The settings for CW will likely not be good for SSB. With NR on, the desired signal should be enhanced, but really weak signals may be at or near the noise level, and may not appear as a valid signal, giving the appearance that weak signals may be covered up and not detected. You should normally run with the NR turned off and only use it when warranted.

Lyle (KK7P) provided the following information:

The K3 NR algorithm is a trade-off between noise reduction and stability.

The wider the dynamic range of signals presented to it, the more aggressive it must be, which in turn makes it less useful for applying small amounts of noise reduction.

The NR pays some attention to the settings of AGC Slope and AGC Threshold. The higher the Threshold, and the steeper the Slope, the more aggressive NR must be at its lowest settings to prevent instability (SCREECH!!!!). We have chosen to make the NR somewhat more aggressive in all cases to prevent instability.

Thus, if you have AGC Threshold set low and the Slope set to flat, the NR will be more usable on weaker signals.

In the end, NR is a filter, and like any filter it will affect what is being passed through it. The weaker the signal, the more it appears like noise and the harder it is to separate the two.

Finally, NR is part of a general field of psychoacoustics. What some people like, others barely tolerate. Somewhat like tastes in music...

We are always looking at ways of doing this better, and went through a very lengthy several months ago to improve NR. You can be sure that we will revisit it when we review the status of the K3 firmware and operation.

## Spurious Signal Removal:

In an e-mail written on May 26, 2009, Wayne, N6KR, addressed this with the following.

"The K3 is a very high-dynamic range, down-conversion superhet. It's nearly impossible to remove all birdies in such a design. But we think we did a pretty good job of it :)

If you have any left (that you can hear with an antenna connected), you can "remove" them using the SIG RMV menu entry in the CONFIG menu. Below are the instructions for SIG RMV, from the latest owner's manual (not released yet)."

**SPURIOUS SIGNAL REMOVAL:** Fast-tuning "birdies" can in some cases be eliminated using the SIG RMV menu entry. We recommend that this be attempted only on birdies that can be heard with an antenna connected. (In SSB modes, simply leaving auto-notch on may be effective.)

If another station is interfering with your receiver, TAP the "NTCH" button once to select auto-notch (the "NTCH <>" icon will be displayed on the lower left of the display) and it can remove most/all of the interfering signal. You can turn it on, off and if you HOLD the "NTCH" button, you can manually adjust the size of the NOTCH by turning the VFO B dial.

For Auto Notch to work (only works in SSB), AGC must be turned "On". This feature will find and remove one carrier (sometimes more than one).

Manual Notch removes one carrier at a specified pitch, and can be used in CW and DATA modes as well as SSB.

### STEPS:

(1) Set up the receiver for a desired band, and select the mode you'd normally use in the target band segment. Also set the DSP passband controls to the settings you use most often (e.g., NORMalized by holding NORM).

(2) Locate a birdie to be removed. A birdie is considered "fast-tuning" if a small change in the SHIFT control (e.g. 50 Hz) moves the birdie about 400 Hz or more (with a narrow filter selected, this small shift may move it completely out of the passband). Such birdies result from UHF harmonics of the VFO. If you test a birdie using SHIFT, be

sure to return SHIFT to its normal setting before continuing, because the value of SHIFT affects the birdie frequency.

(3) Change the SIG RMV parameter to 0. This will save necessary information about the birdie, including the present VFO frequency, mode, filter, and SHIFT value.

(4) Try different parameter values, starting with -1 or 1, to see if the birdie can be removed (shifted out of the audible passband). Use the smallest effective value.

Important: If you change the parameter, exit the menu and see if the birdie has disappeared. Each SIG RMV value applies to a single 100-Hz VFO segment, e.g. from 28135.30 to 28135.39, so you'll have to tune slowly through that 100-Hz range using 10-Hz steps to make sure the birdie has been completely removed. You may hear a slight tuning artifact as you tune in and out of the mapped-out segment.

Note: In many cases, you'll need to apply SIG RMV to at least two adjacent 100-Hz VFO segments. Also, if you use more than one filter bandwidth or SHIFT setting in the present operating mode, you'll probably need to map out the birdie more than one time. If you change your CW sidetone pitch, you may need to re-do the procedure.

To undo SIG RMV:

If you tap CLR while you're in the SIG RMV menu entry, all birdie information for the present band will be permanently deleted.

Limitations:

SIG RMV applies only to CW/SSB/DATA modes, and only to the main receiver (at present). Do not remove birdies on transverter bands; instead, remove them on the I.F. band. On each band from 160-6 m, up to 60 birdies can be removed. But as mentioned above, each one may have to be mapped out from adjacent VFO segments as well as for different filter and SHIFT settings. So the number of birdies that can be completely removed is around 5 to 20 per band."

### **Ten-Tec Titan Amplifier Connection:**

Connect a line with an RCA connector from the K3 "Key Out" and an RCA connector to the Titan "Push To Talk/Vox".

This provides RF generation before keying the amp relay and keeps the relay closed until RF ceases.

### **SUB Receiver:**

To hear VFO A in one ear of headphone and VFO B in the other ear, perform the following steps:

Turn [SUB] on by pressing the [SUB] button

Hold the [B SET] button until "b Set" is displayed on the K3 screen

Tap the [ANT] button and select antenna

Tap the [B SET] button

## Frequency Calibration:

My procedure is only a slight modification of Method 2 in the manual. The AGC will keep the audio level nearly constant under all but the worst conditions.

1. Go to the MENU item, CONFIG: VCO MD
  - a. Set the parameter to CAL
    - i. Exit menu to allow the routine to run
  - b. Select fine VFO resolution (1 Hz).
2. Select CW mode, fast AGC, and set bandwidth to 500 Hz. (You don't have to have a CW xtal filter.)
3. If using WWV, do the calibration only when the carrier is unmodulated. If you try it when there are audio tones, you may end up tuning to one of the sidebands.
4. Using CWT, tune to the calibration signal. This will get you very close to zero beat.
5. Locate CONFIG:REF CAL.
6. Tap SPOT to enable the sidetone, and adjust its level to be about the same as that of the signal. You should hear the "beat", a fluctuation in volume. If necessary, adjust the sidetone level for the strongest beat. (If you can't hear it when adjusting the sidetone level, shift the VFO a few Hz.) Tune the VFO for the slowest possible fluctuation, probably less than 1 per second. If you have trouble hearing the fluctuations, try feeding the audio to an analog voltmeter or oscilloscope.
7. Note the VFO display frequency. If it isn't within about 100 Hz of the calibration signal, adjust the REF CAL frequency in small increments, always retuning the VFO for zero beat as above, until the VFO display reads with 100 Hz of the calibration frequency. Once you get within about 100 Hz, set the VFO for exactly the calibration frequency, and adjust REF CAL for zero beat as above. (You may be close enough initially.)
8. Using the K3 utility, save the configuration. (Or record the REF CAL value.)
9. Cancel SPOT and exit the menu.

If you change menu items related to K3 calibration settings, reboot the K3.

## Factory Reset:

### Parameter Initialization

Menu parameters are stored in non-volatile memory (EEPROM and/or FLASH). It is possible, though rare, for parameters to become altered in such a way as to prevent the firmware from running correctly. If you suspect this, you can reinitialize parameters to defaults, then restore a previously-saved configuration (or re-do all configuration steps manually; no test equipment is required).

- If you have a computer available to do configuration save and restore, run the K3 Utility program, then use the Configuration function to save your present firmware configuration.
- If you don't have access to a computer, you should write down your menu parameter settings. The most important are CONFIG:FLx BW and CONFIG:FLx FRQ (for each installed filter, also tap SUB to obtain sub receiver crystal filter settings, if applicable). You should also note the settings of option module enables (all CONFIG menu entries starting with 'K', e.g. CONFIG:KAT3). If you don't record your crystal filter and option settings, you may have to remove the K3's top cover (and sub receiver, if installed) to verify which options as well as crystals filters are installed, as well as the frequency offsets noted on the crystal filters (depends on filter type).
- Turn the K3 OFF (using the K3's POWER switch, not your power supply).
- While holding in the SHIFT/LO knob (which is also labeled NORM below), turn power ON by tapping the K3's POWER switch. Let go of the SHIFT/LO knob after about 2 seconds. You should now see EE INIT on the LCD.
- When EE INIT completes after a few seconds, you may see ERR PL1 or other error messages due to initialization. Tap DISP to clear each message.
- If you have a computer, restore all parameters using the Configuration function of the K3 Utility program.

If the above does not restore properly, disconnect the K3 from the power source for a minute. Run the K3 Utility, bring up the help file, and type force into the search box, then click on the List topics button for a step by step.

## Filter Settings:

Main RX:	2.7	2.1	400 and 200
Sub RX:	2.7	2.1	400

### Use of the filters:

Normal operation in CW:	2.1	Listening on 500 Hz BW or more No need for other filter if band is normal
Contest operation in CW:	400	(also used if heavy QRM)
Normal operation for SSB:	2.7	(this filter is much wider than 2.7 but that's no problem in normal band conditions)
Contest operation in SSB:	2.1	(needed when band is crowded)
For CW & Data Modes	200 Hz,	5-Pole filter recommended by Elecraft

Diversity means one receiver (main) in one ear and the other receiver (sub) in the other ear. Both sub and main filters have the same offsets in diversity mode.

Use the 200 Hz when the going is tough in both CW and Data modes.

FL1	13 kHz	(FM)
FL2	2.7 or 2.8 kHz	(SSB and casual CW/Data)
FL3	1.8 or 1.5 kHz	(Narrow SSB & wide CW/DATA)
FL4 or	400 or 500 Hz 700 kHz	(Normal CW/DATA) (Casual/normal CW/DATA)
FL5 Or	200-350 Hz 350 Hz	(CW/DATA in heavy QRM) (Optimum 45.45 RTTY)

According to Elecraft (Wayne, N6KR) while matched filters are optimal for matching receiver performance, they're not strictly necessary if you're only doing non-diversity receive. The K3 will select an appropriate filter bandwidth for each receiver independently.

He further says, if you want to do diversity receive, you'll need filters with matched offsets. Our 8-pole filters all have offsets of 0.00, but can also do this with 5-pole filters as long as the offsets are matched for a given crystal filter bandwidth. At the time you order the filters, you can request matched sets

### K3 Crystal Filter Setup:

The following text is copied from an e-mail written by Wayne, N6KR on February 13, 2009.

Taken from Offset (FLx FRQ)

-----

There are many considerations that go into the positioning of the K3's crystal filters with respect to the I.F. center frequency. We shift them upward if the filter is too wide to be centered at  $F_c/2 + 200$  Hz, thus preserving the lower edge of the filter around 200 Hz. This usually happens only in CW mode, typically with lower sidetone pitches. The result is optimization for opposite-sideband rejection. You can argue for a different approach, but the K3's success in CW DXing and contesting suggests that this approach is just as valid as any other.

I don't recommend trying to fool the firmware by adjusting the crystal filter offsets; I'd use the marked values. Changing them is likely to cause unwanted side-effects, since the filter passband is inverted for complimentary modes (CW/CW REV, LSB/USB, DATA/DATA-REV). The exception to this rule is when fine-tuning the offsets of 5-pole filters on the main and sub-receivers in order to provide best performance in diversity mode. These adjustments will rarely be larger than +/- 20 Hz anyway.

Bandwidth (FLx BW)

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It's OK (but not necessary) to fudge the bandwidth of specific filters (FLx BW). For example, INRAD's 8-pole "400 Hz" filter can be declared as 0.4, 0.45, or 0.5 kHz, depending on where you want this filter to be kicked in as WIDTH is rotated. The audible effect is subtle. Ed Muns, WOYK, has described this technique in detail elsewhere.

Gain (FLx GN)

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Regarding filter loss compensation: I recommend simply sticking with what's on page 46 of the owner's manual (1 to 2 dB compensation for 400-500 Hz filters, 3 to 4 dB compensation for 200-250 Hz filters, and 0 for all others) unless you have a very specialized application and lots of time on your hands.

There are three reasons for this. (1) The \*perceived\* loss of a crystal filter is a function of both actual loss and S/N ratio. If a narrower filter has a little more loss, you may not notice it. (2) Most signals are flattened slightly by AGC anyway. (3) Adding a lot of gain to filters can again cause side-effects, such as interaction with RX EQ settings and variations in gain between main and sub if the filter complements are different (e.g., during diversity use).

In fact we recently issued new guidelines for factory-assembled K3s: all crystal filters' FLX GN numbers will be to 0 except for 400-500 Hz (1 dB) and 200-250 Hz (2 dB). Customers are free to optimize these further, of course -- but most won't need to.

The K3 firmware does not allow the selection of any filter other than the 2.7 (2.8) kHz filter for CW transmit. The menu does not prohibit setting it to any filter but it will not transmit and will present you with a TXF error.

## Roofing Filters:

There are recommendations in the K3 Owner's manual (see "Crystal Filter Setup", indexed, on page 46) :

Use VFO A to set the gain in dB. In general, you'll want to add 1-2 dB for 400-500 Hz filters, and 3-4 dB for 200-250 Hz filters.

and there is a table (with essentially the same guidance) in K3 Utility Help for the Filter Configuration page. I've lost the table formatting in cut/paste from Help, but here's the information:

Narrow crystal filters tend to have more passband loss than wide filters.

You can compensate for this effect by specifying an amount of added gain to use for each filter in receive mode. Enter the desired gain "boost", in decibels, for each filter. You may specify a value between 0 and 8 dB.

Elecraft recommends using the initial loss compensation values below. You may wish to further adjust the values for your particular filters.

### Part Number Bandwidth Gain

KFL3A-200	0.20 kHz	3-4 dB
KFL3A-250	0.25 kHz	3-4 dB
KFL3A-400	0.40 kHz	1-2 dB
KFL3A-500	0.50 kHz	1-2 dB
KFL3A-1.0K	1.00 kHz	0 dB
KFL3A-1.8K	1.80 kHz	0 dB
KFL3A-2.1K	2.10 kHz	0 dB
KFL3A-2.7K	2.70 kHz	0 dB
KFL3A-2.8K	2.80 kHz	0 dB
KFL3A-6K	6.00 kHz	0 dB
KFL3B-FM	13.00 kHz	0 dB

Drew (AF2Z) makes the following comments:

If I use a 50uv signal generator to set the filter gains for equal readings on my K3's AFV meter I arrive at the following compensations:

2.70 KHz = 0db

500 Hz = 4 db

250 Hz = 8 db

This is way more aggressive than the suggested amounts but these figures do sound equalized to my ears for real signal conditions. The factory settings are noticeably weak. Maybe my particular filters are just not up to par.

K3iNetwork used with Skype will provide streaming audio to your iPhone, iPad or iPod Touch. This is a no cost audio solution for those who are using the K3iNetwork. Review detailed instructions at: [www.ks7d.com/Skype.htm](http://www.ks7d.com/Skype.htm)

This can be accomplished with non-Apple using a PC and MSN Messenger and a public ILS server (or the old MSN Connect) or other desktop sharing utility like Adobe Connect to share desktop control between two PCs. The shack PC needs to be left turned on and connected to the K3 and running an application such as Ham Radio Deluxe. Just share desktops between the two computers with control enabled and you can run any keyboard mode.

## Button and Switch Function List:

### Elecraft K3 Front Panel Controls: Physical Location vs. Functions

Wayne Burdick, N6KR; Revised 2-4-2009

#### Front Panel

Left Side Controls (left of LCD display bezel); "Primary Controls"

Note: The first two switches in this column are "rocker" switches, with each end having different functions.

Column 1 (leftmost controls on panel)

Switch 1, left end: TAP = BAND DN; HOLD = VOX on/off

Switch 1, right end: TAP = BAND UP; HOLD = QSK on/off

Switch 2, left end: TAP = MODE DN; HOLD = ALT mode select

Switch 2, right end: TAP = MODE UP; HOLD = TX TEST on/off

Switch 3: POWER on/off

[TX and Delta F LEDES]

[PHONES JACK]

[MIC JACK]

Column 2

Switch 1: TAP = MAIN MENU; HOLD = CONFIG MENU

Switch 2: TAP = XMIT; HOLD = TUNE

Switch 3: TAP = RX ANT on/off; HOLD = sub ANT switch (MAIN/AUX)

Knob 1: Smaller knob = MAIN AF GAIN; Larger knob = SUB AF GAIN

Knob 2: Smaller knob = MAIN RF GAIN; Larger knob = SUB RF GAIN

Column 3

Switch 1: TAP = DISP (VFO B alt. displays on/off); HOLD = METER (TX bargraph)

Switch 2: TAP = ATU TUNE; HOLD = ATU normal/bypassed

Switch 3: TAP = ANT1/2; HOLD = Assign name to antenna (complex operation)

Center Controls (under LCD display bezel)

Note: Columns 1 and 2 contain two knobs each; they are the

"Multifunction Controls". Functionally, they are grouped left/right:

col 1/knob 1 and col 2/knob 1 are the DSP filter passband controls, and

col 1/knob 2 and col 2/knob 2 are control transmit functions. All

four knobs are left of the VFO A knob.

Column 1

Knob 1: SHIFT or LO-CUT; TAP = SHIFT/LO select; HOLD = NORMalize filter

Knob 2: SPEED or MIC GAIN; TAP = SPEED/MIC select; HOLD = DELAY (QSK/VOX)  
Column 2

Knob 1: HI-CUT or WIDTH; TAP = HI/WIDTH select; HOLD = PRESET I/II select

Knob 2: CMP or PWR; TAP = CMP/PWR select; HOLD = MONITOR (sidetone or voice)

[VFO A knob]

Column 3 (right of VFO A knob)

Switch 1: TAP = FREQ ENTRY; HOLD = SCAN

Switch 2: TAP = FINE on/off (1 Hz); HOLD = COARSE

Switch 3: TAP = RATE (10/50 Hz); HOLD = VFO LOCK

Switch 4: TAP = SUB on/off; HOLD = VFO LINK/UNLINK; LONG HOLD = DIVERSITY

Right Side Controls (right of LCD display bezel)

Columns 1/2/3: Numeric Keypad and Misc. Function Group

Note: The switches in these three columns form a 3x4 keypad for use with FREQ ENTRY, so the switches are listed by \*row\* (three each) in the list below. Switch '.' is used as the decimal point following MHz; switch '<' is used as an 'enter' key to accept the direct numeric frequency input. Directly below this 3x4 switch array is the VFO B knob.

Row 2

Switch 4: TAP = PRE on/off; HOLD = ATTN on/off

Switch 5: TAP = AGC slow/fast; HOLD = AGC on/off

Switch 6: TAP = XFIL; HOLD = DUAL PB (dual-PB CW or dual-peak RTTY filter)

Row 3

Switch 7: TAP = NB on/off; HOLD = NB LEVEL (VFO A = DSP NB, VFO B = IF NB)

Switch 8: TAP = NR on/off; HOLD = NR ADJUST (use VFO B)

Switch 9: TAP = NOTCH; HOLD = MANUAL NOTCH (use VFO to set notch freq.)

Row 4

Switch .: TAP = SPOT (or autospot if CWT is ON); HOLD = PITCH

Switch 0: TAP = CWT on/off; HOLD = TEXT DEC

Switch <: TAP = AFX on/off (simulated stereo, etc.); HOLD = DATA MD

Columns 4/5: Memory, DVR, Message, and RIT/XIT Control Group

Note: The switches in this group form an irregular array, treated differently in the list below. The first row of 2 switches are for memory and DVR control. Below this is a group of four switches arranged in a circle (M1-M2 on the top of the circle, and M3-M4 on the bottom), with a fifth switch in the middle (REC). These are used for instant message play (CW/DATA or with DVR). They

may also be used with the memory switches as per-band "quick memories".

Below the circular group is a group of 3 switches used for RIT/XIT control, and below this, the RIT/XIT OFFSET control knob.

Row 1: Memory and DVR Controls (rightmost controls on panel)

Left switch: TAP = V>M; HOLD = AF RECORD (DVR)

Right switch: TAP = M>V; HOLD = AF PLAY (DVR)

Row 2 (circular switch array):

M1-M2 (top), M3-M4 (bottom): TAP M1-M4 = msg play; HOLD M1-M4 = msg repeat

Center switch: TAP = MSG RECORD (follow w/M1-M4); HOLD = msg bank 1/2 select

Row 3: RIT and XIT Controls

Left switch: TAP = RIT on/off; HOLD = programmable menu function PF1

Right switch: TAP = XIT on/off; HOLD = programmable menu function PF2

Bottom switch: TAP = RIT/XIT CLR (tap again to undo clear, if enabled)

Knob: RIT/XIT offset control

## APF Function:

This feature allows you to hear CW signals much better.

1. You need to set CONFIG:DUAL PB to APF.
2. Now select preset I, and turn APF ON by holding the DUAL PB switch.
3. Then switch to preset II and turn APF off again using the DUAL PB switch.
4. From then on you should be able to alternate between I and II and have APF turn on and off.
5. You can also vary WIDTH for each preset. For example, for preset I, I set WIDTH to 200 Hz and APF on. For preset II I set WIDTH to 400 Hz with APF off.
6. Make sure the CW signal is centered in the passband at your selected pitch, before engaging APF. For very weak signals, rely on your ear to match the pitch, rather than using the K3 SPOT or CWT features.
7. Press the XFIL/DUAL PB key to enable the APF for more than one second.
8. After turning APF on, use the APF SHIFT (now 5 Hz resolution) and/or FINE tuning (1 Hz resolution) to peak the signal.

## CWT Function:

1. CWT tunes to zero-beat your CW pitch
2. The capture range of CWT is about (+/-) 125 Hz
3. If the signal is noisy, CWT will not find it
4. Multiple signals within your passband will confuse CWT
5. CWT will adjust to the pitch set for your sidetone
6. In CW Mode, hold the "PITCH" button and adjust to a comfortable sound for you using the VFO A knob

7. If CWT finds false zero beat at a very different pitch, narrow the filter (width control) to a few hundred Hz and tune until you hear the signal with CWT "On" and view the CWT moving dot display - then tap "SPOT"

### Split Operation

1. Tap the "A/B / SPLIT" button once to place VFO A and VFO B on the same frequency
2. Tap the "A>B / BSET" button twice to place VFO B in the same mode as VFO A with all the same settings
3. Set VFO B to the desired frequency and press and Hold the "A>B / SPLIT" button
4. Hold the "A/B BSET" button to access VFO B
5. Hold the button labeled "REV" on the K3 to listen on your transmit frequency while in Split mode. There is no tap vs. hold with this button, when held, it temporarily reverses VFOs A and B.

The K3 has 11 "bands" that are cycled through with the up/down "Band" switch. Each band permits reception on a range of frequencies and transmission on the ham band located within each of these ranges. While this might be (probably is) published someplace, I empirically determined the following:

Band 1:	0.490 --> 2.999 MHz
Band 2:	3.000 --> 4.799 MHz
Band 3:	4.800 --> 5.999 MHz
Band 4:	6.000 --> 8.999 MHz
Band 5:	9.000 --> 12.999 MHz
Band 6:	13.000 --> 16.999 MHz
Band 7:	17.000 --> 18.999 MHz
Band 8:	19.000 --> 22.999 MHz
Band 9:	23.000 --> 25.999 MHz
Band 10:	26.000 --> 30.000 MHz
Band 11:	44.000 --> 54.000 MHz

When you are on one of these bands and tune a frequency within the given range, it will be remembered the next time you cycle to that band.

If you want to guarantee that you are on the given ham band, program the quick memories and recall one of them. Using the M1-M4 memories you can then remember a

given mode and band segment. I use mine for CW, SSB, RTTY and PSK, each at the "nominal" frequency for the mode.

## Preamp:

Don (W3FPR) comment, "One thing I might point out up front is that the K3 has more gain (sensitivity) than is usable in many situations. Do not turn on the preamp unless it is necessary. If you can hear an increase in the "band noise" when an antenna is connected (with the preamp off), then the preamp will do you no good other than increase the band noise you have to listen to. If you have a lot of "band noise" even with the preamp off, then turning on the attenuator for that band may help too.

Remember the "rule" -- if the band noise increases when the antenna is connected, you have more than enough receiver sensitivity for that particular band and antenna - additional gain will only result in hearing more noise on the band and will decrease the receiver dynamic range (as well as subjecting your ears to a constant local noise level)."

## VOX Delay

VOX "hold time" is set with the DELAY knob on the K3.

HOLD (press for  $\frac{1}{2}$  sec) the SPEED/MIC knob to activate this function. Delay is shown in seconds on the LCD as you adjust it.

See VOICE MODE VOX SETUP in your Owner's manual for more information (Pg 29 in the Rev D manual).

## N3ME Elecraft K3 Menu Settings:

There are two groups of menu type functions: *MAIN* and *CONFIG*.

Tap the *MENU/CONFIG* button to access the *MAIN* menu; hold the *MENU/CONFIG* button to access the *CONFIG* menu.

Use the VFO A dial to select *MAIN* or *CONFIG* options.

Use the VFO B dial to set the option as desired.

Tap the *DISP/METER* button for information about the current setting displayed on the front panel of the K3. For most entries, the default parameter value is shown in parentheses at the start of the help text.

MAIN Menu entries:

Entry	Default	Description
ANTIVOX	000	<p>ANTIVOX prevents the VOX (voice operated transmission) circuit from being activated by the speaker output or room noise from entering the microphone.</p> <p>Increase the default setting to the level where the ambient noise level will not key the microphone.</p> <p>If you wish to use VOX, you need to set "VOX GN" and "ANTIVOX" levels to what works best for the operator.</p>
AFX MD	Delay 5	<p>Audio Effects.</p> <p><b>Delay 1-5</b> (quasi-stereo); <b>BIN</b> (L/R phase shift)</p>
ALARM	OFF	<p>Alarm/auto-power-on time</p> <p>Tap [1] to turn alarm on/off</p> <p>Tap [2] to set <b>HH</b></p> <p>Tap [3] to set <b>MM</b></p>
LCD ADJ	8	<p>LCD viewing angle and contrast.</p> <p>Use higher settings, if the radio is used at or above eye level. If adjusted incorrectly, bar graphs will be too light or too dark during CW keying</p>
LCD BRT	6	<p>LCD backlight brightness.</p> <p>Use <b>DAY</b> in bright sunlight and <b>2</b> to <b>8</b> for indoor lighting.</p>
LED BRT	4	<p>LED brightness (relative to LCD backlight brightness)</p> <p>Exception: if "LCD BRT" is set to <b>DAY</b>, LEDs are set to their maximum brightness.</p>
MIC SEL	FP.L	<p>Microphone/line audio source, mic gain range and mic bias.</p> <p>Source selections are:</p>
		<b>FP.L</b> for front panel 8-pin <b>MIC</b> jack
		<b>RP</b> for rear panel 3.5 mm <b>MIC</b> jack
		<b>LINE IN</b> for rear panel <b>LINE IN</b> jack (monoaural)
		Tap [1] to toggle between <b>.Low</b> and <b>.High</b> MIC gain range
		Tap [2] to turn mic <b>BIAS</b> on/off (turn on for electret mics)

MIC+LIN	ON	<p>If set to <b>ON</b>, and "MIC SEL" is set for <b>FP</b> or <b>RP</b>, the MIC or Line Input can be used for transmit audio.</p> <p>Note: Setting "MIC SEL" to <b>LINE</b> overrides the "MIC+LIN" menu entry and its parameter becomes "N-A".</p> <p>When "MIC+LIN" is in effect, rotating the "MIC" control displays MIC gain. You must set "MIC SEL" to <b>LINE</b> temporarily in order to adjust the LINE IN gain.</p>
MSG RPT	6	<p>Sets interval (in seconds) between message repeats (<b>0 - 255</b>). To repeat a message, hold [M1] - [M4].</p> <p>A 6 to 10 second interval is suggested for casual CQing. Shorter intervals may be needed during contests. A longer interval is suggested for periodic CW beacons.</p>
RPT OFS	0600	<p>Set transmit offset in kHz for repeater operations. Adjustable from <b>0</b> to <b>5000</b> kHz. Store per band and per memory. Use "ALT" to select a +/- offset or simplex operation.</p>
RX EQ 1	-----	<p>Receiver audio graphic equalizer</p> <p>VFO A is used as an 8 band bar graph, where each character displays the boost or cut (<b>-16 dB</b> to <b>+16 dB</b> in 1 dB increments) for a given AF band. The 8 bands are 0.05, 0.1, 0.2, 0.4, 0.8, 1.6, 2.4 and 3.2 kHz.</p> <p>Tap [1] - [8] to select an AF band. VFO A selects boost/cut. Tap [CLR] to reset all bands to +0 dB.</p>
TX EQ	-----	<p>Transmit audio graphic equalizer (voice modes only)</p> <p>Functions the same as the RX EQ and can be adjusted while in transmit mode.</p> <p>"TX*EQ" indicates "TX ESSB" is in effect</p>
TX*EQ		
VOX GN	050	<p>Adjusts the VOX sensitivity to match your microphone and voice</p>

## CONFIG Menu Entries:

Menu entries that include [T] are tech mode entries. These are only visible if "CONFIG:TECH MD" is set to "ON".

Entries described as "Advanced" or "Troubleshooting" should only be changed with caution.

Tap [DISP] to view the default value, which appears in parentheses at the start of the help text.

Entries marked [SUB] have two settings: one for the main receiver and one for the sub receiver. If a sub receiver is installed, the menu entries will change to identify which receiver is being set up by showing **RF** (main receiver) or **SUB** (sub receiver) at the left end of the parameter display. When in the **SUB** mode, the [SUB] icon will flash.

Prior to adjusting sub receiver menu parameters, turn the sub receiver on by tapping [SUB]. Turn "SUB AF" gain up and main "AF" gain down.

Even if the sub receiver is turned on, when you first enter the menu, **RF** will be in effect and the [SUB] icon will be turned off. Tap [SUB] to switch to the sub receiver parameter as required.

2 TONE [T]	OFF	(Troubleshooting) Enables built-in 2-tone generator for SSB transmit tests. The internal 2-tone generator only works if LSB or USB mode is selected. After setting the tone ON, exit the menu and tap "XMIT". You can use [MIC] to adjust the amplitude of one of the tones; the other's amplitude is fixed.
ADC REF [T]	5.00	Allows calibration of the voltage reference used by the K3 to measure and display certain values, such as the rig's supply voltage. (Optional.) First, disconnect anything attached to the ACC jack. Next, locate the <i>ADC REF</i> menu entry. It will initially show 5.00 volts as the reference voltage. Using a DMM set to DC volts, measure the actual voltage at pin 2 of the ACC jack. This must be done while the ADC REF parameter is being displayed. (Note: The (-) probe of the DMM should go to the K3's chassis ground, e.g. at the <i>GROUND</i> lug.) Finally, use VFO A to set the <i>ADC REF</i> menu parameter to what you measured at pin 2.
AF GAIN	HI	Sets the AF gain range. Available selections are <b>HI</b> or <b>LO</b>
AF LIM	Nor 030	(Advanced) Adjustable AF output limiter for use when "AGC" is turned off. This can protect your ears if a large signal appears. Signals or noise above the threshold will sound highly distorted due to the limiting action, reminding you to back down the "AF" or "RF" gain. Typical settings for those who often turn "AGC" off are 17 to 23. Available selections are 001 to 030
AFSK TX	FIL ON	If set to <b>FIL ON</b> , A 400 Hz transmit audio filter is added (via DSP) in order to maximize transmit signal to noise ratio. This may be useful in high noise computer environments. This applies only to AFSK-A mode (RTTY).
AFV TIM [T]	1000	(Advanced) Integration time for <b>AFV</b> and <b>dBV</b> displays in ms. See also VFO B alternate display information.
AGC DCY	Soft	Available selections are Normal and Soft AGC decay characteristic; applies to both fast and slow AGC. <b>NOR</b> selects traditional linear decay. The <b>SOFT</b> setting can reduce IMD caused by traditional AGC, and is especially useful in "pileup" conditions, in some cases making it unnecessary to turn AGC off. Also see <i>AGC HLD</i> , below, which can reduce AGC IMD even further.

AGC HLD	0.20 12/2011  0.05 Original	Available selections are 0.01 up SLOW AGC "hold" time. Specifies the number of seconds that the SLOW AGC level is held after the signal drops. This can be used to reduce IMD caused by traditional AGC, and is especially useful in "pileup" conditions, in some cases making it unnecessary to turn AGC off. Also see <i>AGC DCY</i> , above.
AGC PLS [T]	Off	(Advanced) NOR enables AGC noise pulse rejection.
AGC SLP [T]	000 12/2011  0.07 Original	(Advanced) Higher values result in "flatter" AGC (making signals at all amplitudes closer in AF output level).
AGC THR [T]	008	(Advanced) Sets AGC onset point; a higher number moves the onset up.
AGC-F [T]	200 12/2011  080 Original	(Advanced) Sets slow AGC decay rate; a higher number means faster decay.
AGC-S [T]	020	(Advanced) Sets slow AGC decay rate; a higher number means faster decay.
AUTOINF [T]	NOR	(Advanced) If set to <b>AUTO 1</b> , the K3 will send band data on its RS232 port for use with devices such as the SteppIR™ antenna on every band change. (Note: This setting may not be compatible with PC software applications that use the "AI" remote control command.)
BAT MIN	11.0	Low-battery warning threshold; <b>11.0</b> recommended. (This refers to a battery used as the K3's DC power supply, not to the 3-V backup battery for the real-time clock.) If the voltage drops below this level, the operator will be alerted with a <b>BAT LOW</b> message. The menu parameter flashes if this occurs within the menu, so the level can be easily tested.
BND MAP	{band} In	Allows you to remove one or more bands from the <b>BAND</b> rotation. Use <b>BAND</b> up/down to select bands, then set them to <b>In</b> or <b>Out</b> using VFO A. (Works with transverter bands, too.) You can still get to mapped-out bands using memory recall, direct frequency entry, or computer-control commands.

CW IAMB	A	Iambic Keying Mode ( <b>A</b> or <b>B</b> ) Mode B is more efficient for operators who use "squeeze" keying (pressing both paddles at once)
CW PADL	Tip=dot	Specifies whether left keyer paddle is <b>Dot</b> or <b>Dash</b> .

CW QRQ	OFF	
CW WGHT	1.10	<p>CW keying weight Adjusts element/space timing ratio for the internal keyer</p> <p>Tap [1] to select SSB + CW (allows CW keying in SSB modes) or SSB - CW (no CW keying in SSB modes (this is the default)</p> <p>Tap [2] to specify how the '@' character should behave when embedded in remote control KY ("key") packets. Select "@" = STOP to allow the '@' character to terminate KY - packet transmission (this is the default) Select '@' = AC to have '@' translated into its Morse equivalent</p> <p>Tap [3] to select OLD or NEW QSK (default). NEW QSK reduces keying artifacts in the presence of QRN or QRM. OLD mutes/unmutes slightly faster.</p> <p>Tap 4 to select <b>VOX NOR</b> or <b>AUTO OFF</b> for CW VOX (hit-the-key CW). <b>VOX NOR</b> is the default, making the CW VOX setting permanent. The <b>AUTO OFF</b> setting turns CW VOX off on power-up. This is useful with some station setups where a computer might accidentally key the K3 at power-up.</p>
DATE	N/A	<p>The real time clock date, shown in the format selected by CONFIG:DATE MD (MM.DD.YY or DD.MM.YY). Tap [1] / [2] / [3] to select month / day / year</p>
DATE MD	US	Select <b>US</b> (MM>DD>YY) or <b>EU</b> (DD.MM.YY) date format
DDS FRQ [T]	{DDS freq}	(Troubleshooting) Controls DDS tuning directly to check DDS XFIL range for synthesizer troubleshooting purposes. Rotate VFO A CCW and CW to find limits where L (lock) changes to U (unlock). Correct DDS frequency is restored after exiting the

		menu and rotating either VFO.



FL1 FRQ [SUB]	rF -0.90	Crystal filter FLx center freq as offset from nominal (8215.0 kHz). Use the offset value specified on the crystal filter's label or PC board. If an Elecraft 5-pole, 200 mHz filter were labeled "-0.91", adjust VFO A for -0.91
FL1 GN [SUB]	0 dB	Crystal filter FLx loss compensation in dB. <b>0 dB</b> recommended for wide filters, <b>2 dB</b> for 400 or 500 Hz filters, and <b>4 dB</b> for 200 or 250 Hz filters
FL1 ON [SUB]	rF YES per mode	Use to specify which filters are available during receive. Each filter must be set to <b>ON</b> or <b>OFF</b> in each mode. Tap [MODE] to set each mode filters. CW = YES DATA = YES AM = YES FM = NO USB = YES
FLT# {mode}	FL1 all modes	Use to specify which crystal filter to use during TX {md} = CW/SSB/AM/FM Choose filters with bandwidths as follows: SSB 2.7 or 2.8 kHz (also applies to data) CW 2.7 or 2.8 kHz AM 6 kHz FM 12 kHz or higher The FM filter, if installed, must be in FL1. Note: If you use a 2.7 kHz 5-pole filter for SSB transmit, you can optionally fine-tune its FLx FRQ parameter to equalize LSB/USB transmit characteristics. Monitor your signal on a separate receiver, using headphones.
FM DEV	5.0	(Advanced) FM deviation in kHz
FM MODE	OFF	If set to OFF, FM will be removed from the mode selections
FP TEMP	40C	Used to calibrate the front panel temperature sensor. This must be calibrated, if you wish to use the REF xxC menu entry to calibrate the optional 1 PPM TCXO. You must convert F to C in order to enter the value. Degree C = (degree F - 32) * 0.555

FSK POL	1	0 = inverted DSK transmit data polarity 1 = normal data polarity
FW REVS	uC 04.12	Rotate VFO A to see firmware revisions: (uC) = MCU (d1) = Main DSP (d2) = Aux DSP (only if KRX3 is installed) FL = flash parameters dr = (only if KDVR3 controller is installed)
KAT3	Auto	KAT3 Antenna Tuner mode Normally set to <b>BYP</b> or <b>AUTO</b> (you can alternate using the ATU switch. Modes <b>L1-L8</b> , <b>C1-C8</b> and <b>Ct</b> are used to test KAT3 relays. Mode <b>LCSET</b> allows manual adjustment of L/C/net settings. When in this MODE, tapping [ATU TUNE] shows the L & C value; C is changed with VFO A. L is changed with VFO B. and [ANT] toggles between Ca and Ct. Tapping [CLR] within this menu entry clears the stored LC data for the current band.
KBPF3 [SUB]	Not Inst	If KBPF3 is installed, set to <b>NOR</b> , exit menu, turn power off/on
KDVR3	Nor	If KDVR3 is installed, set to <b>NOR</b> , exit menu, turn power off/on
KIO3	Nor	Determines function of BAND0-3 outputs on ACC connector
KNB3 [SUB]	rf Nor	(Troubleshooting) The K3 can't be used without a KNB3; The <b>Not Inst</b> setting is for troubleshooting only.
KPA3	PA nor	Set to <b>PA nor</b> if the 100 watt amp is installed. Set to <b>PAIO nor</b> if the 100 watt amp is not installed. Set to <b>PA BYP</b> to disable KPA3 if installed. PA fan test settings: <b>PA FN1-FN4</b> or <b>PAIO FN1-FN4</b> (if transition board is installed, but not the KPA3 module, this setting can be used to test the high power bypass relay).

KRC2	--	<p>Controls the KRC2 band decoder's accessory output settings. Shows <b>ACC OFF</b> or <b>ACC1-3</b> if a KRC2 is detected and "—" if not.</p> <p>To ensure compatibility with both old and new KRC2 firmware, two different 6 meter band decodes are provided.</p> <p>Tap [1] to select <b>BAND6=B6</b> (addr=10) or <b>BAND6=B10</b> (addr=9). Refer to the KRC2 manual for further details.</p>
KRX3	Ant=bnc	<p>If the KRX3 option (sub receiver) is installed, set the parameter to match your selected sub receiver AUX RF source: <b>ANT=ATU</b> (the KAT3's non-transmit antenna) or <b>ANT=BNC</b> (the AUX RF BNC jack on the rear panel). Turn power off, then back on.</p>
KXV3	nor	<p>If the KXV3 option is installed, set to <b>NOR</b>, exit menu and turn power off/on. This option provides RX ANT IN/OUT jacks, low-level transverter I/O (XVTR IN/OUT), and a buffered I.F. output.</p> <p>If the KXV3 is set to <b>TEST</b>, the K3 will use low power (0.10 to 1.50 mW) on all bands, including HF and transverter bands. RF input/output is via the XVTR IN/OUT jacks in this case. Used for troubleshooting. Note: To access the TEST setting, KXV3 must first be set to <b>NOR</b>, then K3 power off/on.</p>
LCD TST	OFF	<p>Changing the parameter turns on all LCD segments.</p>
LIN OUT	Nor 010	<p>Sets the LINE OUT level. LINE OUT connections go to computer soundcard inputs. Settings above 10 may result in overdrive of the soundcard or saturation of the KIO3's isolation transformers. Monitor signals using the computer to avoid this.</p> <p>Note: Normally, "LIN OUT" sets a fixed level receive only output for main/sub (L/R), compatible with digital modes. Tapping the [1] switches "LIN OUT" to = <b>PHONES</b>, where the line outputs match headphone audio, audio level is controlled by AF/SUB gain controls, and both RX and TX audio are available.</p>
MACRO 1	Function	

MEM 0—9	Nor	<p>Can also select "Band Sel"</p> <p>If set to <b>BAND SEL</b>, frequency memories 0-9 ("quick memories") will act like band switches. Once this mode has been selected, you can use <b>V M</b> to assign individual bands to keypad switches 0-9. For example, use [<b>BAND</b>] to get to 160 m, tap <b>V M</b>, then tap <b>#1</b> . From then on, tapping <b>M V</b> , then <b>#1</b> , will take you to your last-used frequency on 160 m, just as if you had used the [<b>BAND</b>] switch. Up to 10 bands can be assigned to keypad switches in this way, including transverter bands. The behavior of memories 10-99 and the per-band quick memories ([<b>M1</b>]-[<b>M4</b>]) will be unchanged.</p>
MIC BTN	OFF	<p>Set to <b>ON</b> if you use a microphone with UP/DOWN buttons compatible with the K3's front panel "MIC" jack. This is not applicable to the Elecraft MH2 or MD2 microphones. "MIC" FUNCTION button is not presently supported.</p> <p>Tapping "UP" or "DOWN" once will move the VFO up or down one step (based on the current tuning rate). Holding "UP" or "Down" will move up or down continuously. If you see the frequency moving up or down continuously, your microphone is not compatible and "MIC BTN" must be set to <b>OFF</b>.</p>
L—MIX—R	A b	
NB SAVE	NO	<p>Set to <b>YES</b> to save noise blanker on/off state per band. Noise blanker levels, both DSP and I.F. are always saved per band regardless of this setting.</p>
PA TEMP	36C	<p>If a KPA3 (100 watt PA module) is installed, this shows the KPA3 heatsink temperature and allows it to be adjusted. The calibration procedure is explained in the manual.</p> <p>If you're operating at high power from a battery, and voltage is dropping enough to cause an erroneous <b>HI TEMP</b> indication, tap [1] to select <b>R ONLY</b> (receive only) temperature sensing, rather than the default (<b>T AND R</b>).</p>
PB CTRL	Shift=.05	

PTT-KEY	rts-dtr	<p>(Advanced) This allows the selection of <b>RTS</b> or <b>DTR</b> RS232 lines to activate PTT or key the K3.</p> <p>Note: If the computer or other device asserts "RTS" or "DTR" while you are in this menu entry, the K3 will switch to <b>TEST</b> mode (zero power output) as a precaution. The <b>TX</b> icon will flash as a reminder. To avoid this, make sure software applications have flow control and/or keying options turned "OFF" while you change the "PTT-KEY" selection.</p>
PTT RLS	20	<p>(Advanced) This provides a delay between release of PTT and dropping of the transmit carrier; intended for use with fast turn around data protocols such as <b>AMTOR</b> and <b>PacTOR</b>. This has no effect on <b>CW</b>, <b>FSK D</b>, or <b>PSK D</b> modes.</p> <p>A value of 20 or higher may be needed to ensure accurate data transmission with these protocols. If sync data or <b>-S</b> is in effect (See <b>SYNC DT</b>), a lower value, typically 10 to 12, is optimal. Also see <b>AMTOR/PacTOR</b> in the manual.</p>
PWR SET	Nor	<p>If set to <b>NOR</b>, the power level on each band follows the present setting of the PWR control. If set to <b>PER-BAND</b>, the power level is saved on each band. This is especially useful with external amplifiers that have varying per band gain, as well as with transverters.</p>
REF CAL or REF xxC [T]	49.379.861	<p>Used to precisely calibrate the K3's reference oscillator. VFO A is used to set the reference oscillator frequency in Hz. There should never be a need to set "REF CAL" outside a range of 49377.000 to 49383.000. Typically it will end up much closer to 49380.000.</p> <p>Tap [1] to alternate between "REF CAL" (Method 1 or 2) and "REF xxC" (Method 3).</p> <p>Tap [2] or [3] to move the data entry point up or down. See calibration procedure in the manual.</p>
RFI DET	Nor	<p><b>NOR</b> enables detection of high RFI at the K3's antenna in receive mode (see <b>HI RFI</b> warning, Troubleshooting). Set to <b>OFF</b> to disable the warning.</p>



SIG RMV [T]	NOR	<p>(Advanced) SPURIOUS SIGNAL REMOVAL: Fast-tuning "birdies" can in some cases be eliminated using this menu entry. We recommend that this be attempted only on birdies that can be heard with an antenna connected. (In SSB mode, simply leaving auto-notch on may be effective.)</p> <p>STEPS: (1) Set up the receiver for a desired band, and select the mode you'd normally use in the target band segment. Also set the DSP passband controls to the settings you use most often (e.g., NORMalized by holding <b>NORM</b>). (2) Locate a birdie to be removed. A birdie is considered "fast-tuning" if a small change in the SHIFT control (e.g. 50 Hz) moves the birdie about 400 Hz or more (with a narrow filter selected, this small shift may move it completely out of the passband). Such birdies result from UHF harmonics of the VFO. <i>If you test a birdie using SHIFT, be sure to return SHIFT to its normal setting before continuing, because the value of SHIFT affects the birdie frequency.</i> (3) Change the SIG RMV parameter to <b>0</b>. This will save necessary information about the birdie, including the present VFO frequency, mode, filter, and SHIFT value. (4) Try different parameter values, starting with <b>-1</b> or <b>1</b>, to see if the birdie can be removed (shifted out of the audible passband). Use the smallest effective value.</p> <p>Important: If you change the parameter, exit the menu and see if the birdie has disappeared. Each SIG RMV value applies to a single 100-Hz VFO segment, e.g. from 28135.30 to 28135.39, so you'll have to tune slowly through that 100-Hz range using 10-Hz steps to make sure the birdie has been completely removed. You may hear a slight tuning artifact as you tune in and out of the mapped-out segment. Note: In most cases, you'll need to apply SIG RMV to at least two adjacent 100-Hz VFO segments. Also, if you use more than one filter bandwidth or SHIFT setting in the present operating mode, you'll probably need to map out the birdie more than one time. If you change your CW sidetone pitch, you may need to re-do the procedure.</p> <p>To undo SIG RMV: If you tap <b>CLR</b> while you're in the SIG RMV</p>
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		<p>menu entry entry, all birdie information for the present band will be permanently deleted.</p> <p>Limitations: <i>SIG RMV</i> applies only to <i>CW/SSB/DATA</i> modes, and only to the main receiver (at present). Do not remove birdies on transverter bands; instead, remove them on the I.F. band. On each band from 160-6 m, up to 60 birdies can be removed. But as mentioned above, each one may have to be mapped out from adjacent VFO segments as well as for different filter and <i>SHIFT</i> settings. So the number of birdies that can be completely removed is around 5 to 20 per band.</p>
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SMTR OF [SUB]	024	S-Meter offset. See calibration procedure.
SMTR SC [SUB]	014	S-Meter scale; S-9 = 50 uV, S-3 = 1 uV with Preamp = ON, and AGC ON. See Calibration procedure.
SMTR PK	OFF	Set to <b>ON</b> for peak reading S-meter.
SMTR MD	Nor	(Advanced) S-meter mode: When set to <b>NOR</b> , preamp/attenuator settings will affect the S-meter. (The default values of <i>SMTR OF</i> and <i>SMTR SC</i> apply to <b>NOR</b> .) If set to <b>ABS</b> , the S-meter reading will stay fairly constant with different preamp/attenuator settings, but <i>SMTR OF</i> and <i>SMTR SC</i> must be carefully realigned for both main and sub receivers.
SPLT SV	NO	If set to <b>YES</b> , [SPLIT], [RIT], and [XIT] on/off states are saved per band.
SPKRS	2	Set to <b>2</b> if using two external speakers. This enables binaural effects in conjunction with the <b>AFX</b> switch, as well as stereo dual-receive, if the sub receiver is installed.
SPKR+PH	Yes	<b>YES</b> = Speaker is On, even when headphones are plugged into the PHONES jack.
SQ MAIN	00	This sets the main receiver squelch value ( <b>0 - 29</b> ).  If VFO A is rotated fully clockwise, the parameter changes to = <b>SUB POT</b> . Squelch for both main and sub receivers will then be controlled by the "SUB RF/SQL" knob, and both main and sub RF gain will be controlled by the "MAIN RF/SQL" knob.
SQ SUB	00	Sets the sub receiver squelch value ( <b>0 - 29</b> ).  If "SQ MAIN" is set to = <b>SUB POT</b> , then "SQ SUB" will also change to = <b>SUB POT</b> . Squelch for the sub receiver will then be controlled by the "SUB RF/SQL" knob, and both main and sub RF gain will be controlled by the "MAIN RF/SQL" knob

SUB AF	Nor	<p>If set to <b>BALANCE</b>, then the <b>SUB AF GAIN</b> control becomes a main/sub AF balance control when the sub receiver is turned on (including diversity mode). In this case <b>MAIN AF GAIN</b> controls the AF gain level for both receivers. When <b>SUB AF</b> is at 12 o' clock, both receivers will be at full volume (main left, sub right). If <b>SUB AF</b> is rotated fully counter-clockwise, you'll hear only the main receiver. If it's rotated fully clockwise, you'll hear only the sub receiver. At intermediate settings you'll hear both. A balance control is very useful for contesting and split operation. But it can also save a lot of AF gain control adjustment (i.e., matching main and sub), since <b>MAIN AF</b> controls both main and sub receiver audio.</p>
SW TEST [T]	Off	<p>(Troubleshooting) To turn on switch test, rotate <b>VFO A</b> until the parameter becomes <b>SCN ADC</b>. You can then hold any switch to see its scan row as well as its column ADC reading. You can also rotate any of the four potentiometers to see their associated ADC readings (main/sub AF gain and main/sub RF gain). If the <b>SUB RF</b> pot is mapped to main/sub squelch, you must switch to <b>FM</b> mode to see its readings.</p>
SW TONE	On	<p>Sets audible control annunciation (tones or audio Morse code) to "On"</p> <p>If set to <b>ON</b>, enables audible switch feedback tones. (Note: For <i>voice</i> feedback on switch press, you may wish to use our <b>K3 Voice</b> program for the PC.) Tones generated: In general, a <i>low-to-high</i> tone pair is generated when a switch function is turned on, and <i>high-to-low</i> when it is turned off. Some switches do not generate tones because they might interfere with received or transmitted audio (e.g. <b>REV</b>, and <b>M1 -M4</b> message play). If <b>CONFIG:RIT CLR</b> is set to <b>UNDO ON</b>, tapping <b>CLR</b> a second time (RIT clear "un-do") produces a unique low-to-high tone pair.</p>

SYNC DT	Function	(Advanced) When "SYCN DT" (sync data) is activated in either SSB or DATA modes, T/R switching times are reduced to optimize for modes such as AMTOR and PacTOR. The "-S" icon turns on. <b>Do not use "SYNC DT" for normal SSB/DATA communications.</b> This cannot be changed within the menu; assign to a programmable function switch. Also see "CONFIG:PTT RLS" (PTT release delay).
TECH MD	On	Set <b>ON</b> to enable <b>TECH MODE</b> menu entries (Those marked with [T] in this menu list.
TIME	##.##.##	Set or view the real time clock. Tap [1] / [2] / [3] to set HH / MM / SS. To see the time and other displays during normal operation, tap [DISP].
TTY LTR	Function	Performs a RTTY FIGS to LTRS shift when the text decoder is enabled in RTTY modes. This cannot be changed within the menu as it must be assigned to a programmable function switch.
TUN PWR	10	If set to <b>NOR</b> , [TUNE] power level follows the POWER knob. Otherwise, it establishes a fixed power lever for [TUNE], overriding the present POWER knob setting. Note 1: "TUN PWR" does not pertain to [ATU TUNE], which always uses 5 or 10 watts and is internally controlled. Note 2: see "CONFIG:PWR SET" for per band power control.
TX ALC [T]	ON	(Troubleshooting) Set to <b>OFF</b> to disable both internal and external transmit ALC (overrides "EXT ALC" setting. Used when adjusting band pass filters in TX mode, or for troubleshooting. Set parameter to <b>ON</b> during normal operation.
TX DLY	Nor 008	(Advanced) For use with external amplifiers that have slow relays; sets the time from "KEY OUT" jack (active low) to first RF in 1-ms steps. To minimize loss of QSK speed, use the shortest delay that works with your amp. Most will work with the default (minimum) setting of 8 ms.
TX DVR	Nor	
TX ESSB	OFF 3.0	(Advanced) Extended SSB transmit bandwidth (3.0, 3.5, 4.0 kHz, etc.) or <b>OFF</b> . Tap [1] to turn on/off, and rotate VFO A to select the bandwidth. If you turn ESSB on/off frequently, you might want to assign this menu entry to a programmable function switch

		(PF1, PF2, or M1-M4).
TX GATE	OFF 00	(Advanced) The TX noise gate can be used to suppress transmitted audio below a certain level, e.g. that of an amplifier fan. Tap <b>1</b> within the <i>TX GATE</i> menu entry to turn the noise gate on/off. Use VFO A to set the desired threshold. Since there's no visual indication that transmit audio is below the threshold, you should adjust it using the transmit voice monitor (MON), ideally while using headphones. Set the threshold high enough to cut off transmit audio due to local noise, but not so high that it causes your voice to drop out too frequently. The <i>TX GATE</i> threshold can be adjusted in <b>TX TEST</b> mode.
TX INH [T]	OFF	(Advanced) If set to <b>LO=INH</b> or <b>HI=INH</b> , an external logic signal can inhibit transmit. When transmit is inhibited, the TX LCD icon flashes.
TXGN HP	14	(Troubleshooting.) Shows transmit gain constant for the present band and power mode, where {pwr} = <b>LP</b> (0-12W), <b>HP</b> (13-120W), or <b>MW</b> (0.1-1.5 mW). The gain constant is updated whenever the <b>TUNE</b> function is activated on a given band <i>at one of three specific power levels: 5.0 W, 50 W, and 1.00 milliwatt. See transmit gain calibration procedure.</i>  On 80 m with high power (> 13 W) selected, you should see <b>PR80</b> as part of the <i>TXGN</i> parameter display. This indicates that the preamp is turned on during QRO transmit on 80 m, and is the default. It should only be turned off for troubleshooting purposes; this is done by tapping [ <b>PRE</b> ].  If <i>TX ALC</i> (above) is <b>OFF</b> , the <i>TXGN</i> parameter can be set manually, at very fine resolution. This should only be done for troubleshooting purposes.
TXG VCE [T]	0.0 dB	(Advanced) Balances voice transmit peak power in relation to CW peak power in [ <b>TUNE</b> ] mode. Typically set between -1.5 to 1.5 dB.

VCO MD [T] [SUB]	Nor 023 (varies with VFO frequency)	(Troubleshooting) VCO L-C range view/change/calibrate. Once the VCO is calibrated (pg. 49), the parameter which appears here will include <b>NOR</b> . You can change the setting to troubleshoot VCO L-C ranges. When finished, set the parameter back to <b>NOR 127</b> , then exit the menu and change bands to restore the original setting. Note1: In this menu entry only, the main/sub receiver prefix ( <b>RF</b> or <b>SUB</b> ) is not displayed at all times. However, the <b>SUB</b> icon will flash as usual when <b>SUB</b> is tapped. Note 2: Tapping <b>8</b> before starting a VCO CAL will force it to use "manufacturing" PLL voltage tolerances ( <b>TOL=MFG</b> is flashed on VFO B). These are more stringent than normal. This might be useful for those using the K3 over a wider temperature range.
VFO B→ A	Function	Copies VFO B's frequency to VFO A. This cannot be used within the menu; must be assigned to a programmable function switch.
VFO CRS	CW = 1.0 SSB = 0.1 Data = 0.1 AM = 1.0 FM = 5.0	Per mode coarse tuning rate (hold [COARSE] and tune VFO A or B. This also applies to the "RIT/XIT" tuning knob if "CONFIG:VFO OFS" is set to <b>ON</b> , and both RIT and XIT are turned OFF.
VFO CTS	400	VFO counts per turn ( <b>100</b> , <b>200</b> , or <b>400</b> ). Smaller values result in easier fine tuning of VFO; larger values result in faster tuning. This does not apply to the [COARSE] tuning rate, which always uses 100 counts per turn.
VFO FST	50	Specifies the faster of the two VFO tuning rates [ <b>RATE</b> ]. The faster rate is <b>50</b> Hz per step by default, but can be set to <b>20</b> Hz if desired. In this case, <i>VFO CTS</i> = <b>400</b> is recommended to ensure adequate fast-QSY speed.
VFO IND	No	If set to <b>YES</b> , VFO B can be set to a different band than VFO A, which allows listening to two bands at once (main/sub). See pg. 37 for independent main/sub band considerations. Note: This menu entry is not available unless the sub receiver is installed. (see "CONFIG:KRX3")

VFO OFS	OFF	If <b>ON</b> , the RIT/XIT offset control can be used to tune VFO A in large steps when both <b>[RIT]</b> and <b>[XIT]</b> are turned off. The step sizes vary with mode (see <i>VFO CRS</i> ), and are the same as the <b>[COARSE]</b> VFO tuning rates.
WMTR HP {PWR} [T]	050	Wattmeter calibration parameter. {pwr} is the power mode: <b>LP</b> (0-12W), <b>HP</b> (13-120W), or <b>MW</b> (0.1-1.5 mW). See calibration procedure for additional information.
The following CONFIG menu items establish parameters for the XV transverter:		
XV1 ON	No	Set to <b>YES</b> to turn on transverter band x (1-9); tap [1] - [9] to select xvtr band
XVx RF	144	Lower edge for xvtr band x (1-9); <b>0-24999</b> MHz. Tap [1] - [9] to select xvtr band.
XVI IF	28	Specifies the K3 band to use as the I.F. for transverter band x (1-9) . Tap [1] - [9] to select the transverter band. I.F. selections include <b>7, 14, 21, 28, and 50</b> MHz.
XVI PWR	H .01	Sets upper limit on power level for XVTR band x. Tap [1] - [9] to select xvtr band. <b>H x.x</b> (High power level) specifies a value in watts, and use of the main antenna jack(s). This should be used with caution, as you could damage a transverter left connected to these antenna jacks accidentally. <b>L x.xx</b> (Low power level) species a value in milliwatts, which requires the KXV3 option. (If <i>CONFIG:PWR SET</i> is set to <b>PER-BAND</b> , the K3 will also save the last-used power setting on each band. This is especially useful for transverter bands.)
XVx OFS	0.00	Offset ( <b>-9.99</b> to <b>+9.99</b> kHz) for transverter band x (1-9). Tap [1] - [9] to select xvtr band. Compensates for oscillator/multiplier chain errors.

XVx ADR	TRN1	<p>This menu entry assigns optional band-decode addresses (<b>TRN1-9</b>, or <b>INT. TRN1</b>) to each transverter band (<math>x = 1-9</math>). Addresses <b>TRN1-9</b> are used to automatically select external transverters, while address <b>INT. TRN1</b> is used to select the Elecraft K144XV internal 2-meter transverter.</p> <p>Addresses <b>TRN1-TRN9</b> can be sent to the BAND DATA outputs on the KIO3 for use with third-party transverters (BAND0-3; see pg. 19 and <i>CONFIG:KIO3</i>). Addresses <b>TRN1-TRN7</b> are compatible with Elecraft XV-series transverters and the Elecraft KRC2 band decoder, so they are sent on the "AUXBUS" serial control line (TRN8 and TRN9 are sent as 0 on the auxBus). <b>INT. TRN1</b> is sent as address 0 to both the BAND DATA lines and the AUXBUS.</p> <p>Tap <b>[1] - [9]</b> to select the desired transverter band display, then rotate VFO A to select the desired decode address.</p> <p>Examples: To configure an Elecraft K144XV internal 2 m transverter as XVTR band 1, set XV1 RF to 144 MHz, XV1 IF to 50 MHz, and XV1 ADR to INT. TRN1. To configure an Elecraft XV432 external transverter as XVTR band 2 , set up XV2 RF as 432 MHz, XV2 IF as 28 MHz, and XV2 ADR as TRN1 (or whatever address matches the DIP switch selection inside the transverter).</p>

## MCU Boot Loader Troubleshooting:

When MCU firmware load is interrupted, or the power-on firmware checksum test fails, the K3 enters its permanent bootstrap loader. You can also force the K3 to enter its permanent bootstrap loader by turning the K3 on with a 10-second press of the power on switch. The MCU boot loader displays "MCU LD" on the K3 front panel LCD. LCD backlighting is normally off in this condition, so the screen appears dark in dim light.

In order to load K3 firmware in this state, follow these steps in sequence.

Turn off your K3 by switching off its power supply or removing the 12V power connector from the back panel.

The K3's MCU can continue to run at very low voltage, and it's important that it start from a power-on reset. For this reason, it may be necessary to wait 30 seconds or so after removing power before power is turned on again.

If you have a P3 panadaptor, remove its power plug so that the P3 does not power on, even for a moment, when you subsequently apply K3 power. Current P3 firmware sends commands during P3 initialization that disturb the K3 MCU boot loader. A future P3 firmware version will remove this constraint, but for now, unplug the P3 power if you find yourself in this situation.

If the K3 Utility is running, select the Port tab, and then close the K3 Utility, using the File Exit menu item or select the red X at the top right of the main window. When the K3 Utility is started the next time, the Port tab provides the clearest indication of a successful connection.

The following two connection options have been suggested - select the one, which works best for your installation:

1. If you are using a sharing program or device between the host personal computer and your K3's RS-232 port (e.g. N8LP's LP-Bridge or microHAM's Router), change the cabling so that the host PC is connected directly to the K3 without an intervening software or hardware "forwarder".
2. Use one serial cable from computer to P3 and another serial cable from the P3 to the K3.

These programs send K3 "programmer's reference" commands, and a K3 in its bootstrap loader isn't yet able to process those commands. Some of the commands might put the K3 bootstrap loader into an unresponsive state.

Restart (reboot) the personal computer that will run the K3 Utility

This isn't often necessary, but occasionally a serial port or USB device driver doesn't properly close when commanded to do so. Rebooting reinitializes the device and its driver.

Restore K3 power and press the K3 power button. You should expect to see "MCU LD" in an otherwise dark LCD, and the red Tx light should be blinking.

Start the K3 Utility and navigate to its Port tab.

The K3 Utility should find the K3 at at 38,400 bps. If the Port tab is selected, you should see a K3 Status popup that looks something like this:

(picture can't be pasted into a straight text message)

If you don't see this status dialog, try pressing Test Communications. If you don't see this dialog, the K3 Utility is not receiving responses to K3 commands, and it cannot load firmware without responses. Information in USB to Serial Adapter Troubleshooting may help to resolve this issue. The problem might be serial port cabling, an incorrect COM port choice, a problem with the Personal Computer, or the K3's internal KIO3 Interface.

Once you see this dialog, press the "OK" button on the K3 Status dialog to continue.

If the View Menu Trace menu item is checked, firmware load can take quite a long time. Uncheck Trace unless you require this diagnostic information.

Select the Firmware tab, and click Send All New Firmware to K3.

The K3 Utility should send MCU firmware to your K3, followed by any other "new" firmware required.

## JT65 Mode Notes:

I use Data A Mode directly from the line in/out to my pc sound card for JT65 and all digital modes with no issues at all.

The K3 is a GREAT rig for JT65, because it needs no external interface. It is the best radio I have ever used for JT65 because of the very flat passband - I can easily see callers plus and minus 1000 Hz.

I have tried the Data Mode, but usually use the USB position because I like the wide passband on receive better. I use the widest receive filter settings possible (to avoid introducing attenuation on the weak signals) and let JT65 do the filtering at audio levels. I don't use any equalization at all on XMIT, and keep my compression and mic ALC settings exactly as when I am on SSB. I realize this is different from PSK31 users, but JT65 is very different from PSK31. Other details on how I set up the radio and software for JT65 EME are here:

<http://www.bigskyspaces.com/w7gj/JT65.pdf>

I use my K3 on EME DXpeditions, and am leaving next week for 5W on a 6m EME DXpedition:

<http://www.bigskyspaces.com/w7gj/Samoa2011.htm>

I hope there will be some new K3 users who will at least listen for me on 6m JT65A mode at the end of this month during their moonset! GL and VY 73, Lance

Well, as I recall, the DATA A passband was narrower than the SSB mode - at least in my K3. I have only 1.8, 6.0 and the standard 2.7 kHz filters installed in mine. When I am calling CQ on JT65 mode on EME, I like to have the widest possible flat passband so I can copy callers who spread out to call me without interfering with each other.

I continue to be puzzled by this thread. Given that WSJT (all modes) is WRITTEN for use on USB by Joe Taylor, WHY would we be interested in utilizing any K3 mode but USB?

I have operated 441 and JT65 utilizing my K3 off the Moon, tropo and MS with no desire to fuss with any other K3 "mode". The K3 works great on USB ... end of story.

Here, USB, no equalization, 2.7Khz filter (5 pole), no compression. M Audio Delta 44 Soundcard, XP dual core machine and no "audio" interface and a homebrew Opto-isolator for PTT

"DATA A" \*is\* USB with no equalisation, etc - that's the point. It allows you to easily switch between SSB and data modes without having to adjust audio processing for each.

I just looked at my passband, with spectran, in DATA A and USB modes with the bandwidth cranked up to 4.0kHz (in both cases) with a 2.7kHz filter. Indeed, the passband seems to be cut off at about 200Hz on the low end in DATA A mode, but not in SSB mode. What's that about?

<http://www.dseven.org/ar/passband.png>

There we are ... IF we wish to run flat and no compression ... DATA A it seems is a natural. I think Lance 7GJ said he ran WITH equalization yesterday, so he would use SSB. Seems like a users preference.

Here, it is very quick to hit pre-programed PF1 allowing a switch to rear panel audio and IF I have used compression I will roll it back and I'm off and running. (I have been known to keep the compression in on occasion)

I never use any equalization at all here, so that is not an issue in my operation ... The K3 runs CW or digital 95% of the time EXCEPT for 6m Es. IF I had to mess with equalization resets, it would become a bit like a dog chasing it's tail at some point. Then I would be looking at DATA mode I bet.

I sensed a bit of aggravation in the responses in that past, that some of us "insist" on using SSB when the Data "modes" were built into the software to simplify our lives. I know when I wrote software and someone avoided one of my programed features, I was "miffed" too.

According to Lyle Johnson, when using PSK operation using the internal PSK32 decoder - always tighten the passband selectivity to 150 Hz or less. You only want to hear one signal at a time.

Don (W3FPR) writes, "With a laptop, controlling the levels may be more complicated, and may require some resistive dividers to work properly. For instance, the laptop headphone output may be greater than what is normally considered "line level" - although if the computer is expecting powered speakers, that one \*may\* be OK. If you have a line out jack from the computer, use that instead of the headphone output.

Connecting the K3 line out to the laptop Mic In, may prove to be troublesome, although some laptops have a software setting that allows either line level or mic level using the same jacks - if that is your case, set it for line level. If your computer is stuck with only mic level input, you may have to reduce the K3 line out signal level by about a factor of 10. A 10k resistor in series with the K3 output line plus a 1k resistor on the computer side of the 10k resistor from the AF line to the AF return (usually the shield). Place that voltage divider close to the computer end to keep noise pickup on that signal line to a minimum.

Remember that the audio input level to the K3 is adjusted differently than the traditional "wisdom" for other transceivers - use TX TEST to set your input level (no RF will be transmitted), use DATA A submode, and adjust the Line In level in the K3 (and/or the output volume and WAV level in the computer) so the ALC meter illuminates the 4th bar with the 5th bar barely blinking. The K3 ALC action does not begin until the audio level hits the 5th bar."