## Using Audio Equalizers to Improve SSB Communications

Bill Leonard

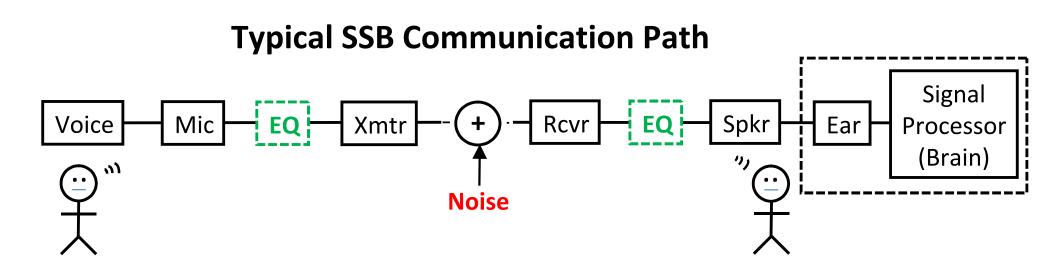
NAØTC - 285 TechConnect Radio Club http://www.naøtc.org/

## **Topics:**

- Commonly Used Methods for Improving HF SSB Comms
- Key points about speech and hearing
- •Equalization:
  - •What is it
  - When to use it
  - Where to use it
    - Transmit applications
    - Receive applications
      - Three cases
  - How to use it
  - What it can do
  - What it cannot do
  - •Some Equalizer Examples:
    - •W2IHY Eight Band Equalizer
    - •MFJ-616/618 Speech Intelligibility Enhancers
    - •A Custom Homebrew Two Band Equalizer

## **How to Improve SSB Communications?**

- 1. Improve Received Signal-to-Noise Ratio (SNR)
- 2. Improve Received Signal-to-Interference Ratio (SIR)
- 3. Use/Improve Signal Processing
  - Digital
  - Analog
    - Equalization



## To Improve Received SNR:

- •Use higher gain *antennas*
- •Raise *peak* transmit power
- •Raise *average* transmit power
  - Common method: compression (RF not audio)
    - •5-10 dB of compression is typical
  - •There is a limit:
    - Must trade-off between distortion and SNR improvement
- "Matched filter" detection
  - Most modern radios have adequate filtering
- •Reduce receiver *noise figure*:
  - Will not help when atmospheric noise is dominant
    - Frequently the case on the low HF bands

## **Signal Processing:**

- Used to improve the extraction of signal information from noise and interference
- •Digital Signal Processing (DSP):
  - "Although God made an analog world, Humans keep trying to digitize it!"
  - "Brick Wall" filters
    - Noise reduction
    - Interference cancelling

Improve SNR/SIR

#### Analog Signal Processing

- •Speech compressors:
  - •RF
- •Better than audio compressors by 8+ dB
- Audio

Increase average transmit power

- •The Human Brain = A very effective signal processor for speech
  - Speech intelligibility
    - •Can be improved by "tailoring" the frequency content of the voice signal to better match the Brain's speech processing algorithm
    - •"Tailoring" <=> Equalization
      - Hams don't normally use equalizers to equalize

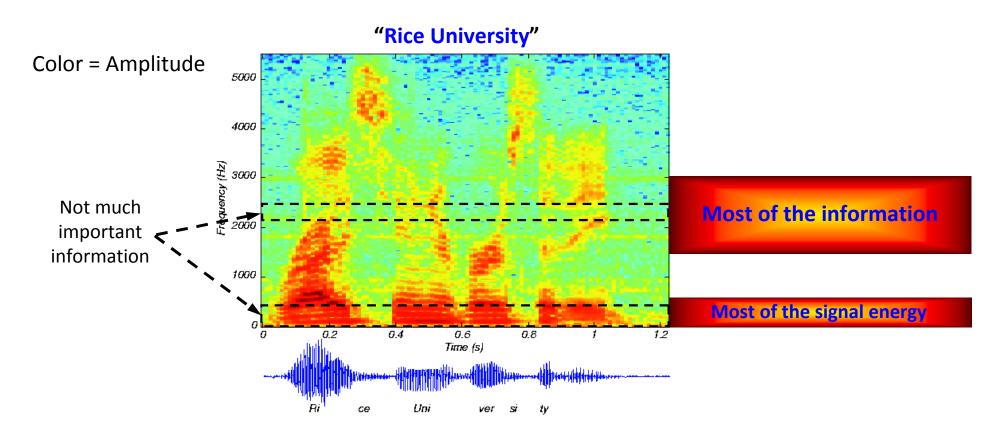
## What is Equalization:

- •In Music applications:
  - "Equalize" the response of two or more signal paths
  - Compensate for deficiencies in the electronics
  - •Emphasize or de-emphasize one or more instruments
- •In Ham applications:
  - •"Equalization" => Tailoring (ie, intentionally distorting) the frequency response to improve the intelligibility of voice signals
    - Most commonly used at the transmit end
      - Emphasize important frequency ranges and de-emphasize others
    - Can also be very beneficial at the receive end
  - Equalized response goals for SSB <u>transmit</u> applications:
    - For rag-chew applications:
      - Improve "fidelity" of the transmit audio for a more natural sounding voice
    - 2) For **DX & Contest** applications:
      - •Increase the "punch" of the transmit audio
      - •Can be fatiguing to listen to for long periods of time
  - Equalized response goals for <u>receive</u> applications vary widely

## What is Important in Speech Recognition:

- •Frequency content of speech varies with time
  - Some frequency ranges are more important than others
- "Good" vs.. "Bad" audio response is very subjective

**Spectrogram**: Amplitude vs. Frequency vs. Time



10/21/2010 7

## What is Important in Speech Recognition (continued):

Speech is made-up of vowels and consonants

#### •Vowels:

- •Longer duration sounds (*Lows*: 30-300 ms)
- •Examples: "ah" & "oh" sounds
- Significant in determining "Who" is speaking
- Contain most of the sound energy in human speech
  - •Energy in consonants can be >20-30 dB below energy in vowels

#### •Consonants:

- •Shorter duration sounds (*Highs*: 10-100 ms)
- •Examples: "f" vs. "s" & "d" vs. "t" sounds
- Significant in determining "What" is being said"
- Contain most of the information
  - •Importance of Bandwidth:
    - Accuracy for single syllables:

```
15.0 KHz Bandwidth => 100%
```

7.0 KHz Bandwidth => 95%

3.3 KHz Bandwidth => 75%

1.0 KHz Bandwidth => <50%

## Where to Equalize?

#### •At Transmitter:

- Best place to put an audio equalizer
  - Point of highest SNR, and
  - Location for most benefit to the station using equalization
  - Settings depend on intended use:
    - •Case 1: Rag-chew
    - •Case 2: DX or contesting
  - •RF & 60 Hz hum getting into transmit audio can be a problem

#### •At Receiver:

- Placed between the receiver and the speaker/headphones
- •Can improve the intelligibility of voice signals when:
  - **Case 1:** They are missing critical frequencies <u>and</u> there is adequate received SNR
  - Case 2: They need more "Punch"
  - **Case 3:** There is a hearing deficiency on the receive end

## **How to Equalize?**

At the TRANSMIT end

Case 1: Improve "fidelity" for rag-chew applications

- •Flat overall response: 100 Hz to 3.0+ KHz
- Sometimes lows are emphasized

**Net** Overall Transmit Frequency Response:



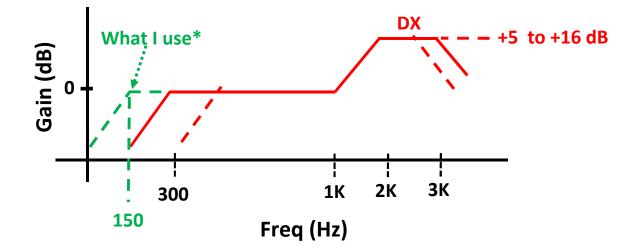
## **How to Equalize?**

At the TRANSMIT end (continued)

Case 2: Increase the "punch" for DX & Contest applications

- •Cutoff "lows": ~300 HZ (typical)
- •Flat "mids"
- •Emphasize "highs"
  - •Starting ~1 KHz
  - •Maximum emphasis from 2 KHz up
  - •Upper cutoff anywhere from 2.4-3 KHz

**Net** Overall Transmit Frequency Response:



<sup>\*</sup>Increases "punch" and maintains fidelity

## **Transmit Equalizer Settings:**

- •Initial settings are based on microphone and rig
  - Available from W2IHY for his equalizers
- •Final settings:
  - Usually arrived at after on-the-air testing
  - Depend on numerous variables:
    - Frequency content of the transmitter operator
    - Frequency response of the microphone
    - •Frequency response through the hardware (transmitter + receiver)
    - Use of compression
    - •Hearing response at the receiving end?
      - •Try to find a station that can use a spectral display to evaluate the audio of received signals
        - •Flex 5000, Perseus, etc
  - The equalizer gain settings usually do <u>not</u> represent the final end-to-end response

## **Equalizers for Transmit Applications**

- •Built-in equalizers
  - Most only have 2-3 sub-bands
  - Some newer high end SDR's have very sophisticated DSP based equalizers
- Heil microphones
  - •HC-4 (DX), HC-5 (rag-chew), HC-6 (all purpose) elements
- •MFJ-655B 8 Band Equalizer + ...
  - Designed for use at transmit end
  - •~\$220 new
  - •eHam: 1.8/5 from 6 reviewers
- •W2IHY 8 Band Equalizer + Noise Gate
  - Designed for use at transmit end
  - •~\$300 new (\$150-200 used)
  - •eHam: **5.0/5 from 174 reviewers**

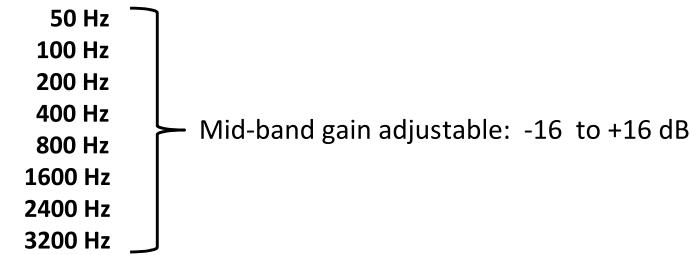




## The W2IHY 8 Band Equalizer + Noise Gate:

#### •8 Band Equalizer:

Splits the input signal into 8 sub-bands with center frequencies of:



#### •Noise Gate:

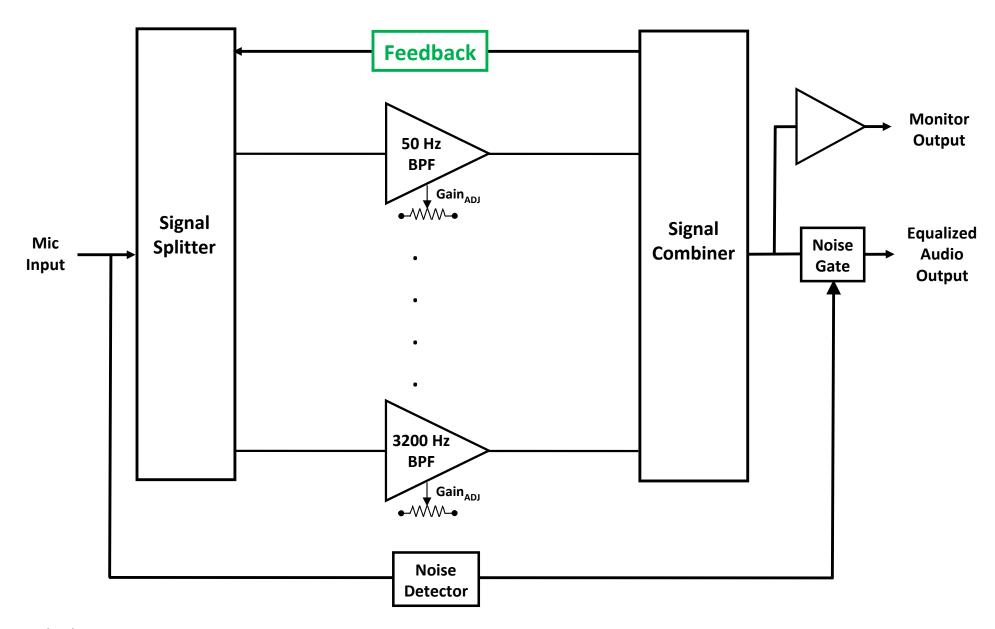
- •Shuts off the audio during periods when there is no speech:
  - •Effectively eliminates the background noise from capturing the transmitter during pauses and between sentences
  - •Does *not* improve communication capability

#### •Monitor:

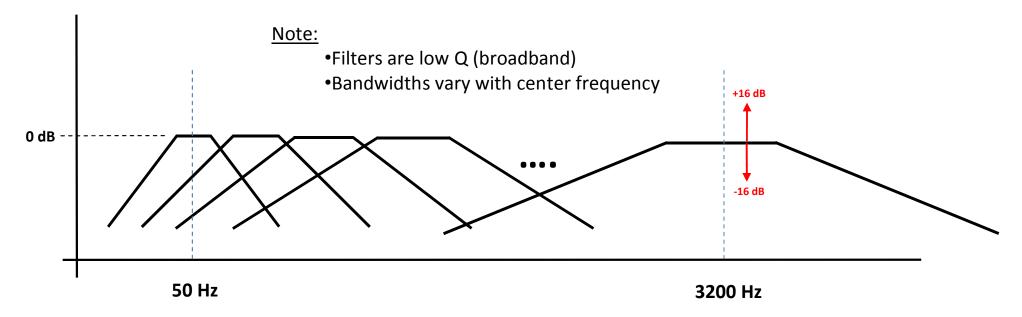
- Somewhat helpful in setting up the equalizer
  - Doesn't account for frequency response of transmitter
- Also allows for use of equalizer on receive

## The W2IHY 8 Band Equalizer (continued):

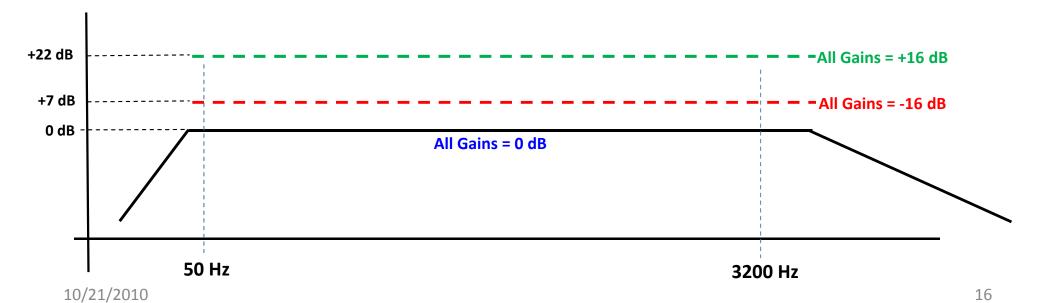
•Uses eight, one-pole bandpass filters (BPF) in parallel



## The W2IHY 8 Band Equalizer (continued):

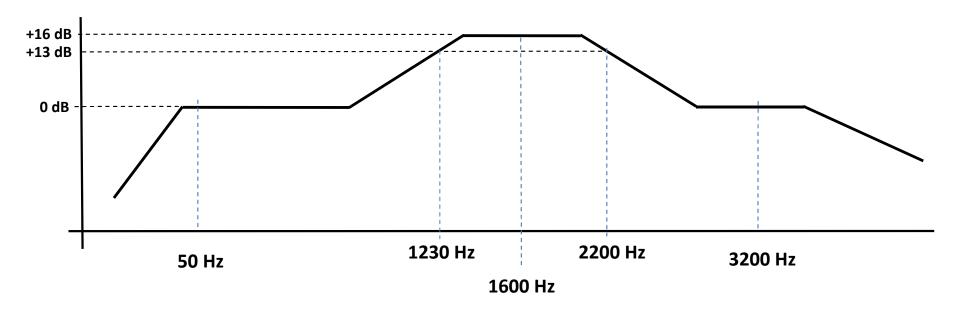


**Note:** Feedback modifies overall behavior of equalizer

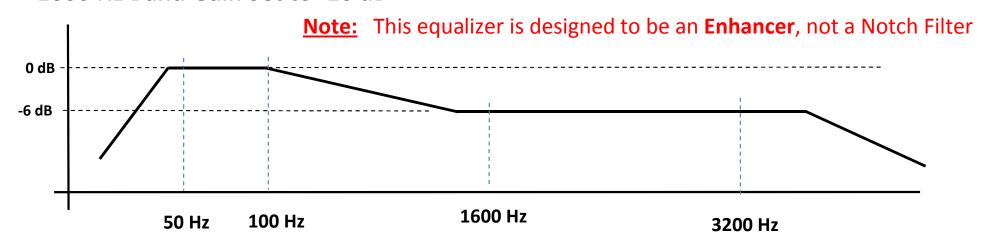


## The W2IHY 8 Band Equalizer (continued):

#### 1600 Hz Band Gain Set to +16 dB

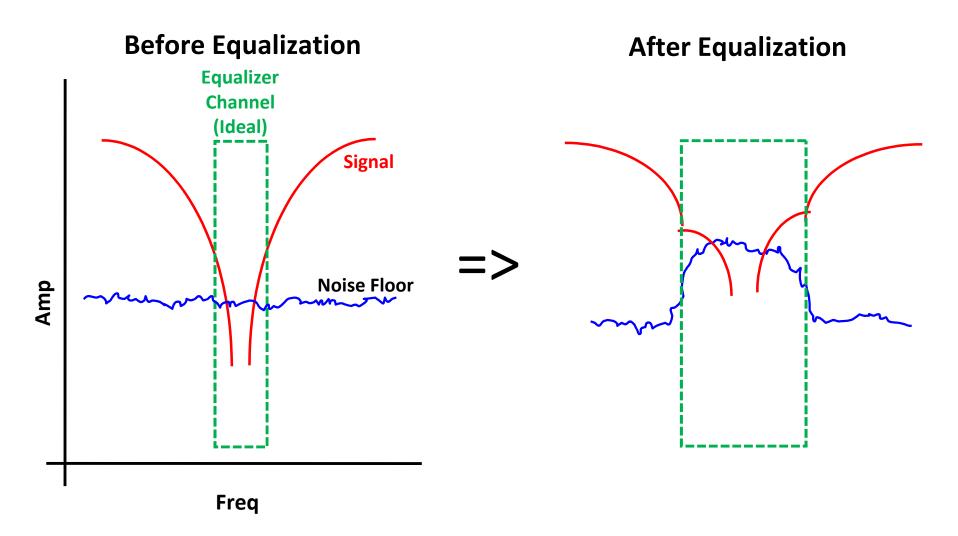


#### 1600 Hz Band Gain Set to -16 dB



## **Signals Below the Noise Floor:**

- •Can <u>not</u> be recovered by an equalizer!
- •Negative SNR + Gain = Negative SNR



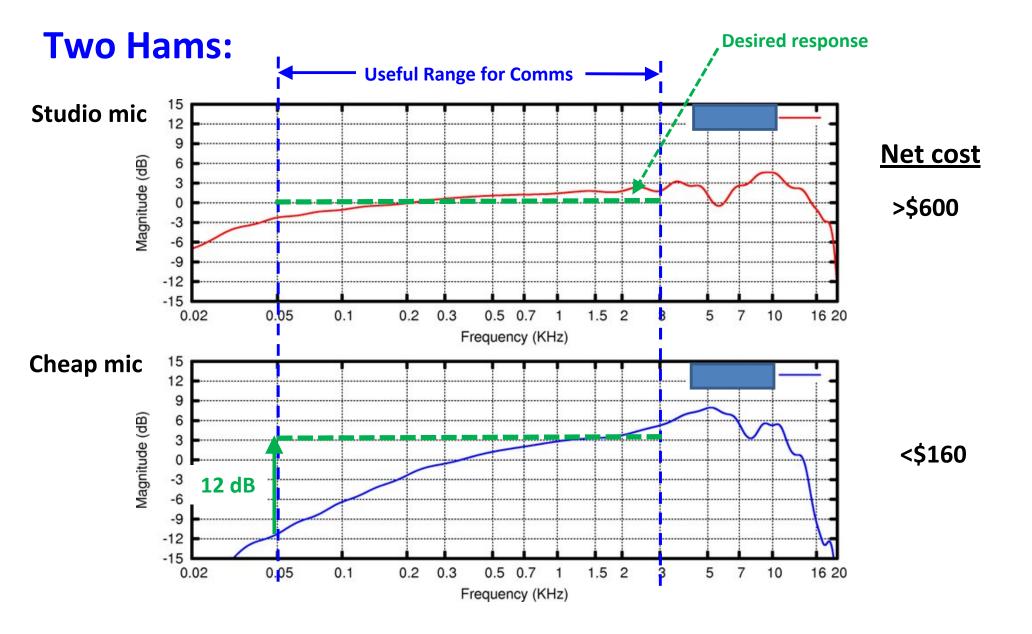
## **Some Equalization Examples:**

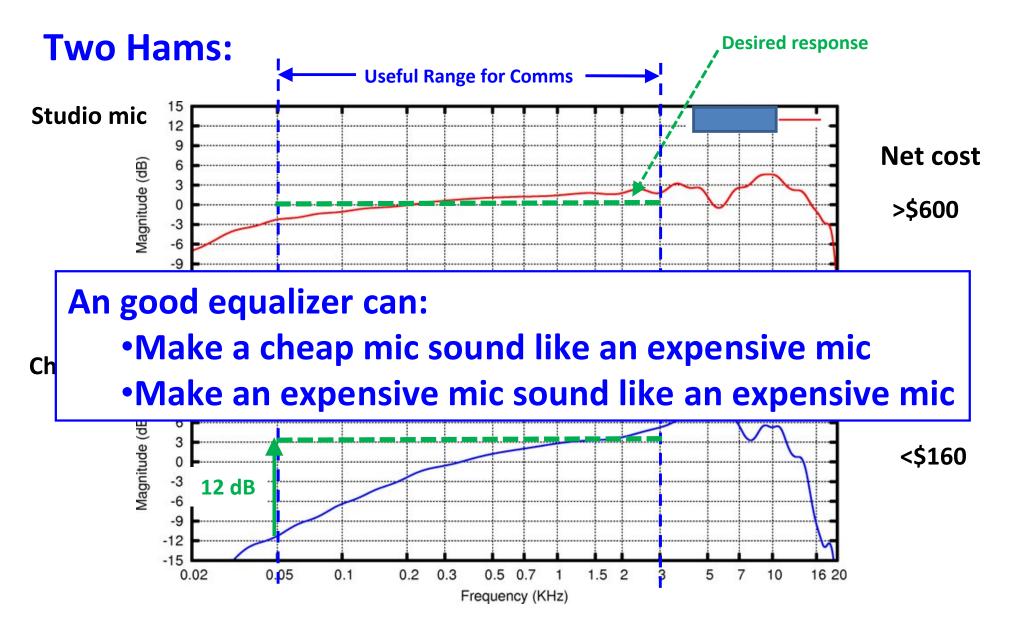
#### **Example 1: Two hams:**

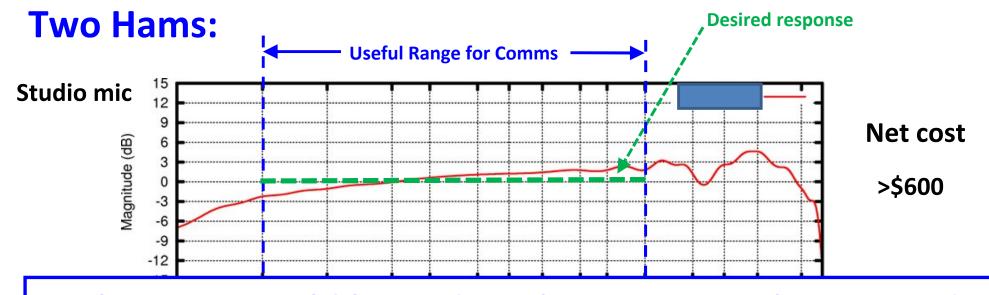
- •Two hams both want flat audio response (ie, for rag-chewing)
- •One buys used cheap mic and a used 8 band equalizer
- •The other buys a new expensive mic and a new 8 band equalizer

### **Example 2: My Station**

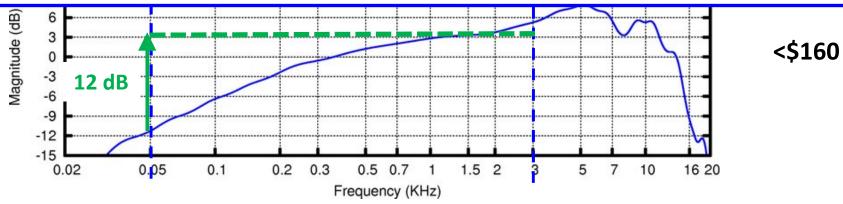
- •lcom 7600
  - •10 dB of compression
  - Bass and treble set at 0 dB
- •W2IHY 8 Band Equalizer
- Heil Proset IC





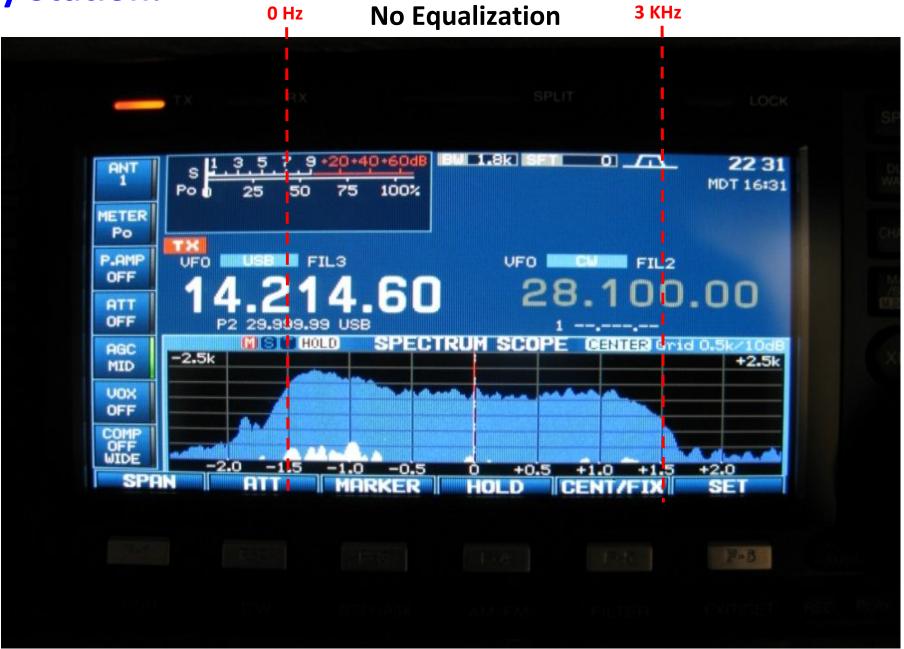


Both Hams would have done better to just buy a good quality communications style microphone for ~\$100!



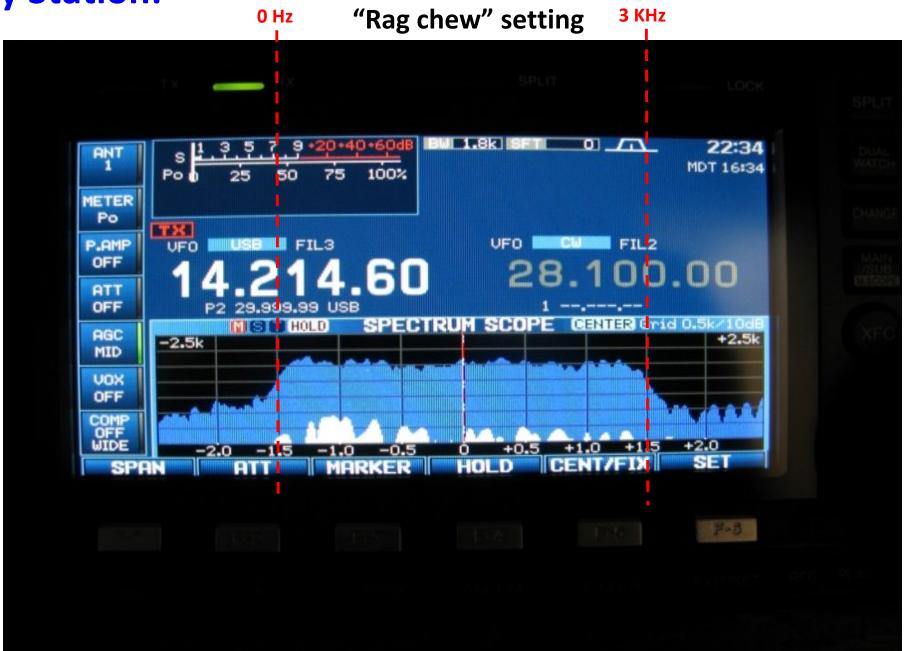
## **My Station:**

## **Transmit Spectrum**



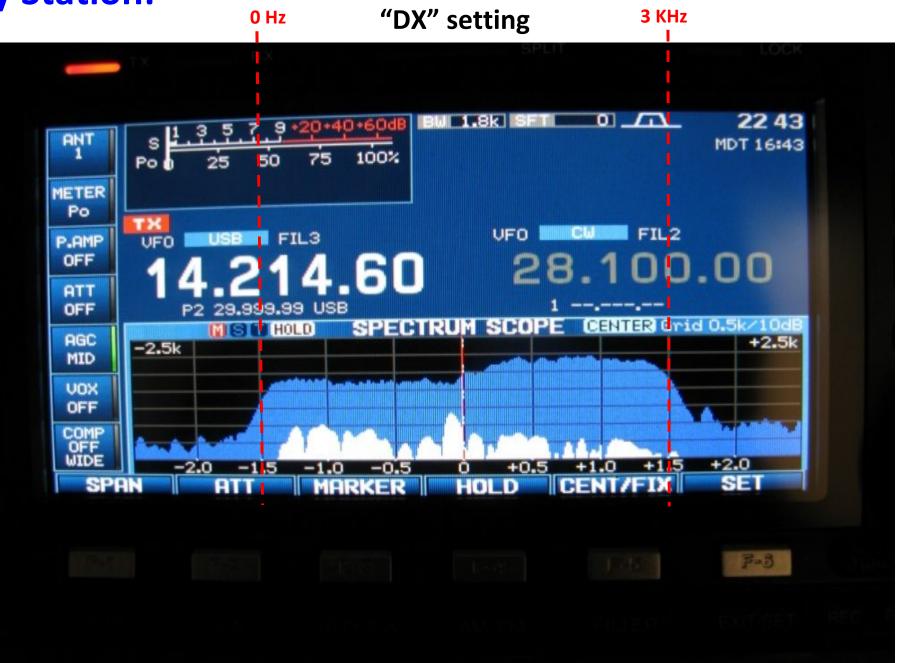
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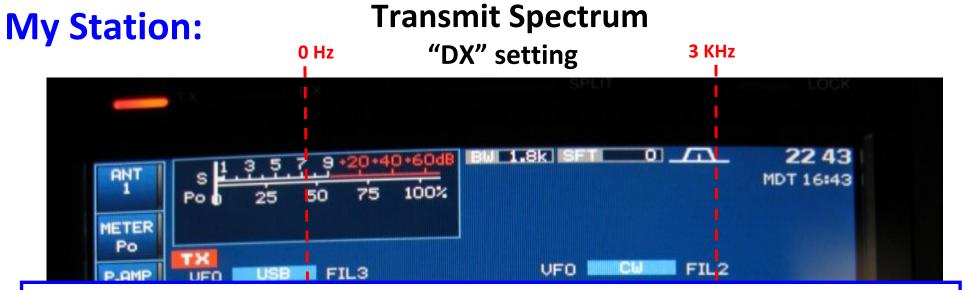
## **Transmit Spectrum**





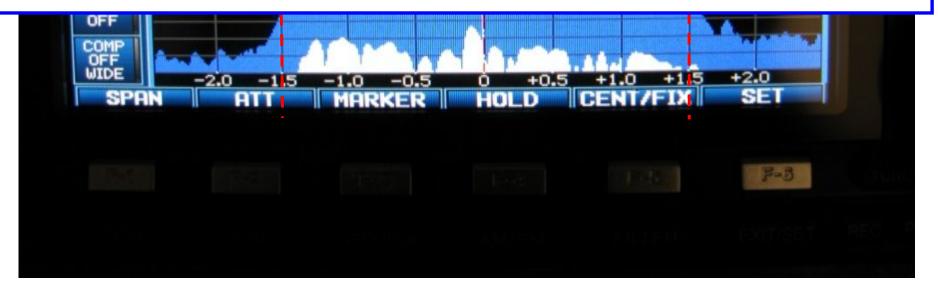
### **Transmit Spectrum**





# Nothing beats a good Spectrum Analyzer/Panadapter for setting up an audio equalizer for transmit!

•It can be used on either the transmit or receive end

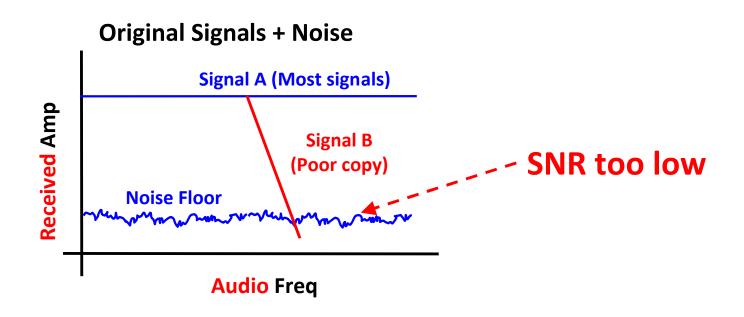


## Equalizers at the Receive end: "Possibly the best kept secret in Ham Radio"

- •How to equalize at the receive end:
  - Some transceivers offer "Bass" & "Treble" adjustments for the receiver
    - Limited capability
      - Only "Highs" and "Lows" can be adjusted
        - Corner frequencies usually not adjustable
      - •May only have +/- 5 dB of gain adjustment range
    - Adequate for many applications
  - •After market units:
    - MFJ-616 Speech Intelligibility Enhancer
      - •4 Band **single** channel equalizer designed for Rx applications
    - MFJ-618 Speech Intelligibility Enhancer
      - •4 Band dual channel equalizer designed for Rx applications
    - The W2IHY 8 Band Equalizer
      - •Designed for transmit applications, but fully functional during receive
  - Custom homebrew design

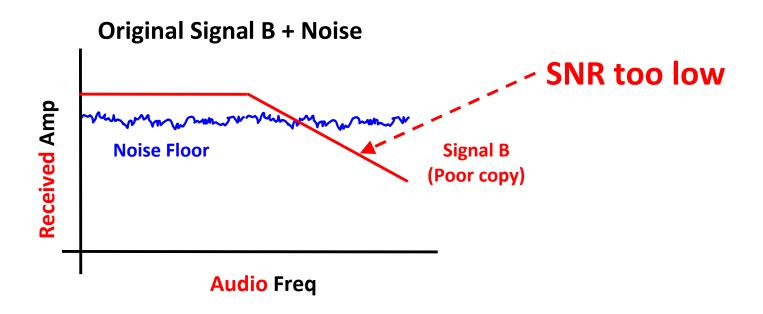
**Case 1:** The transmit signal is missing critical frequencies <u>and</u> the missing frequencies have adequate received SNR:

•Symptoms: <u>some</u> signals are "muffled" or "tinny" and hard to understand, others are not



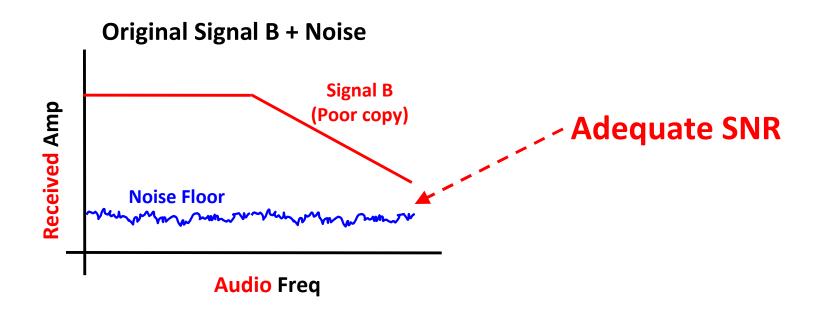
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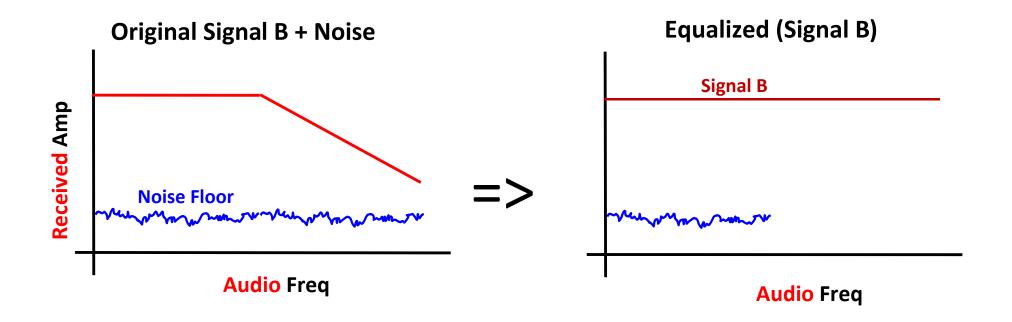
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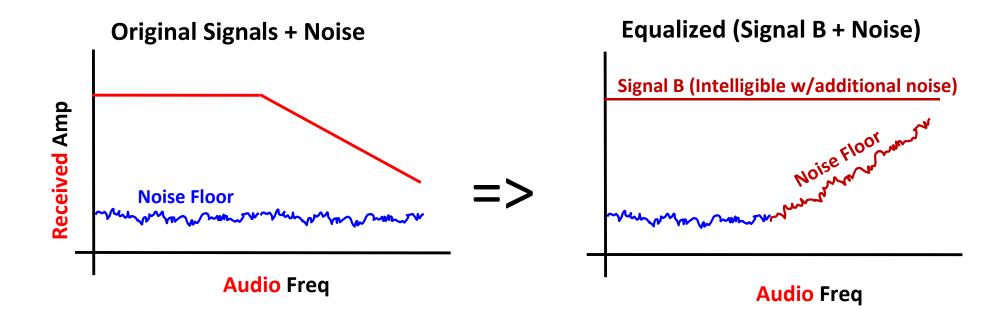
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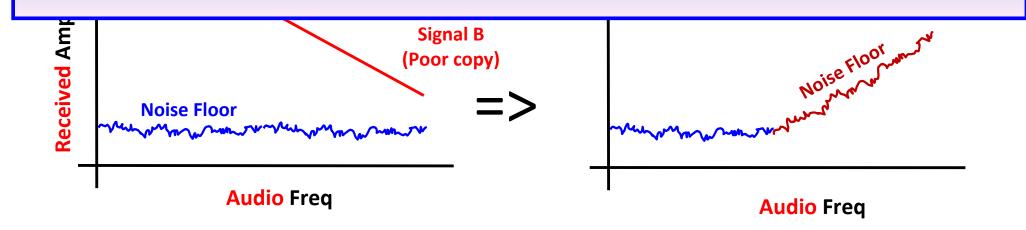
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## How can you get really messed-up audio settings?



## A well intentioned backup OP!



## A well intentioned backup OP!





## A well intentioned backup OP!





# "Keep the damn box covered!"

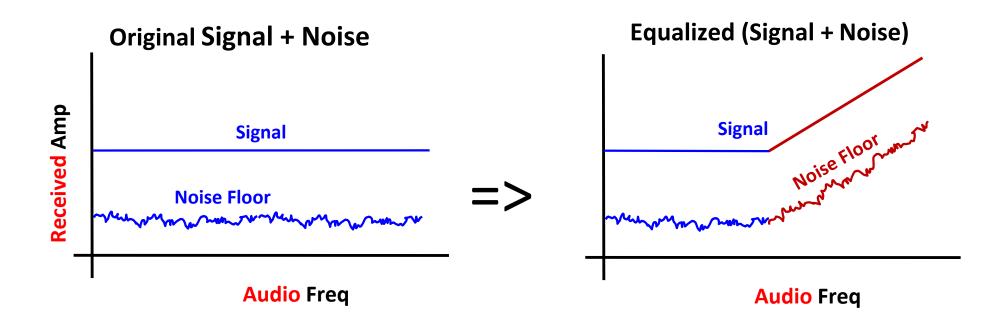


### "Keep the damn box covered!"



Keeping it covered is a good idea even if you don't have any backup ops

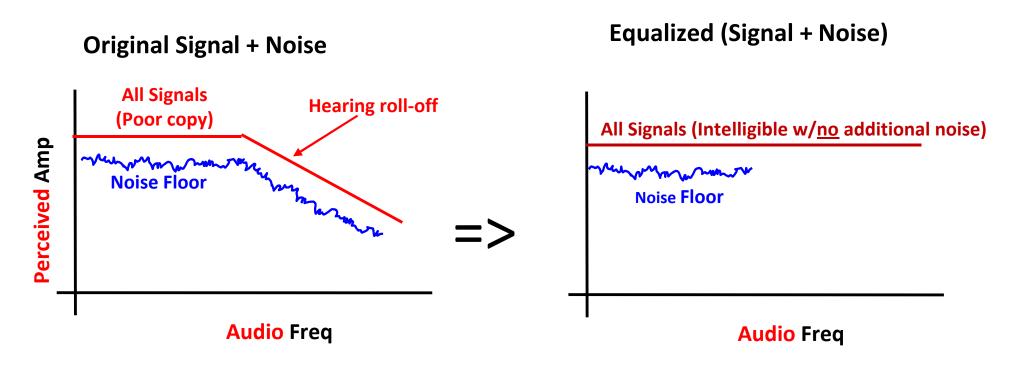
Case 2: The received signal needs more "Punch" <u>and</u> there is adequate SNR at the receiver output:



May be SNR dependent?

**Case 3:** The receive operator has a **hearing deficiency**:

•Symptoms: <u>all</u> signals are "muffled" or "tinny" and hard to understand

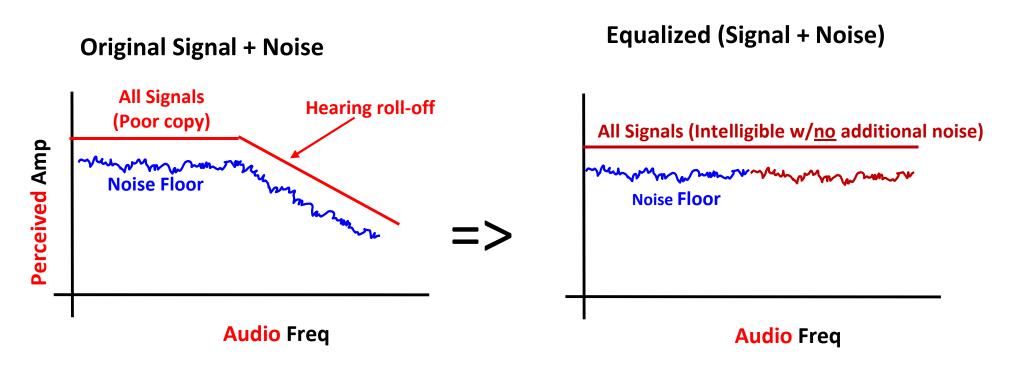


- •Perceived Amp = what the Brain actually hears
- Signal and Noise are <u>equally</u> affected

**Not** SNR dependent

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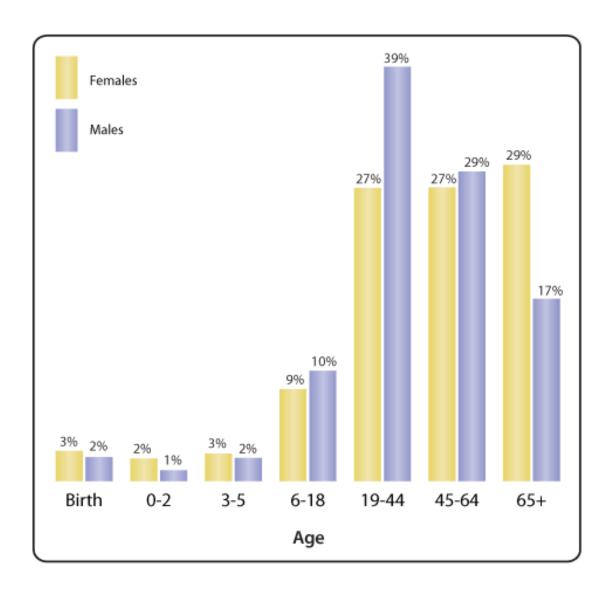


- •Perceived Amp = what the Brain actually hears
- Signal and Noise are <u>equally</u> affected

**Not** SNR dependent

- A hearing deficiency can result from:
  - •Presbycusis: a gradual, age-related reduction in the ability to hear the higher frequency sounds
  - Noise induced hearing loss (NIHL)
  - •Illness induced
- •Age at which hearing loss begins?

## Onset of hearing loss versus age:



### **A Hearing Deficiency Example**

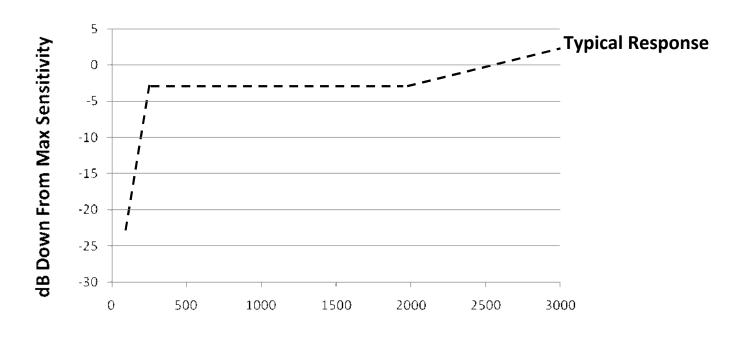
- How is your hearing?
  - •Free On-line hearing test:

http://www.phys.unsw.edu.au/jw/hearing.html

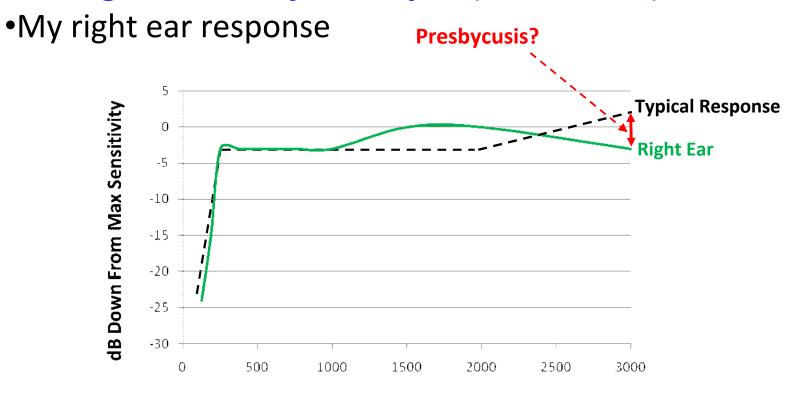
#### Note:

- •Results are <u>relative</u> sensitivities, not absolute
  - •The shape of the response curve can be very accurate
- •Assumes that your sound card & speakers have a flat frequency response
  - •Can use the computer speakers with an ear plug
  - •Good earphones recommended over computer speakers

### Typical response

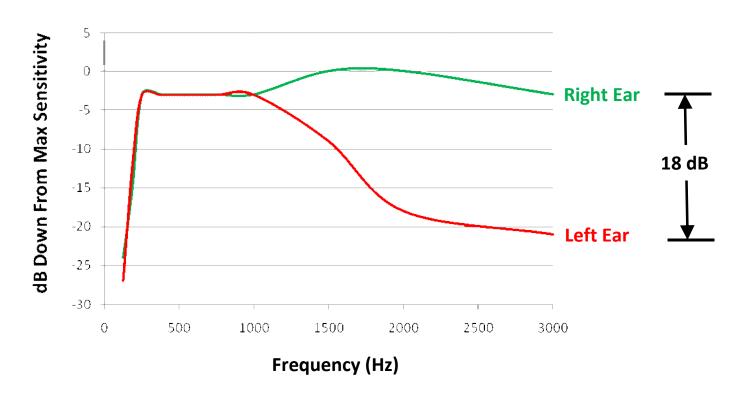


Frequency (Hz)



Frequency (Hz)

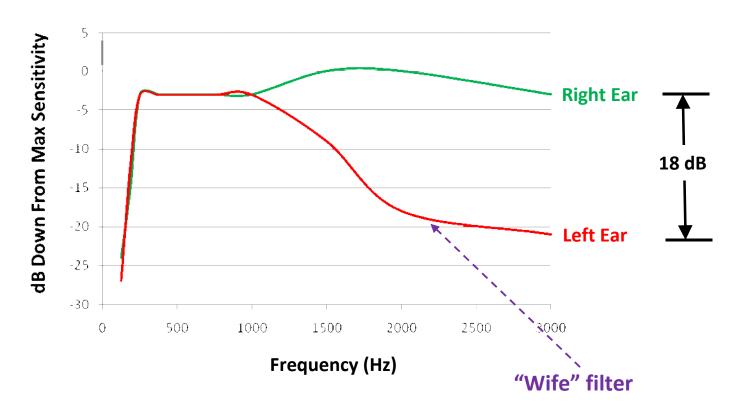
•My left ear has a significant deficiency above 1 KHz



#### **Options:**

- 1. Go monaural (ie, use only right ear)
  - The Brain is programmed for "stereo" reception
  - I copy signals better when both ears are used
- 2. Use an **Equalizer** to "**Equalize**" the response of my left & right ears

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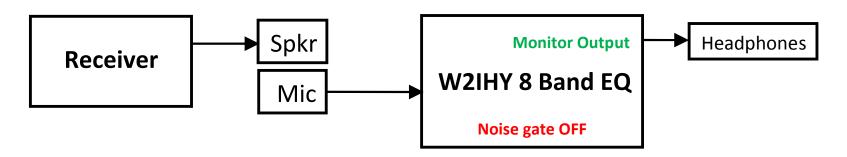


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  - I copy signals better when both ears are used
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### The W2IHY 8 Band Equalizer for Receive:

 Designed for transmit applications, however it is fully functional during receive

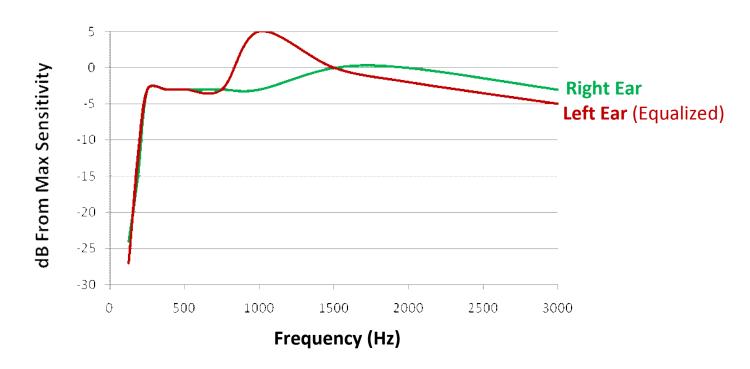


•Settings used to equalize my left ear and right ear responses:

BAND (Hz)	50	100	200	400	800	1600	2400	3200
GAIN (dB)	0	0	0	0	-16	-16	16	16

### **W2IHY 8 Band Equalizer:**

#### **Results:**



This correction resulted is a <u>significant</u> improvement in my ability to copy SSB signals with the left ear!

### **MFJ-616**:

- "Single Channel Speech Intelligibility Enhancer"
  - •4 band audio equalizer *intended for use at the receiver* 
    - Sub-band center frequencies:

300 Hz

600 Hz

1200 Hz

2400 Hz

- Each sub-band gain adjustable +/-12 dB
- Designed to compensate for a wide range of hearing deficiencies
- •\$190 new
- •eHam: 5.0/5 from 6 reviewers



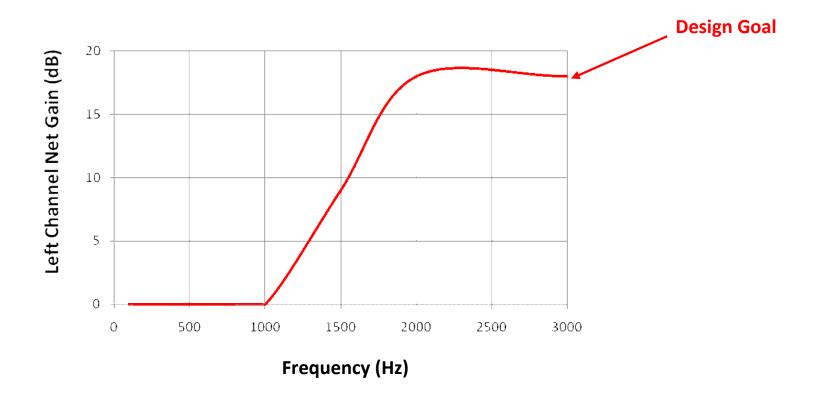
### **MFJ-618**:

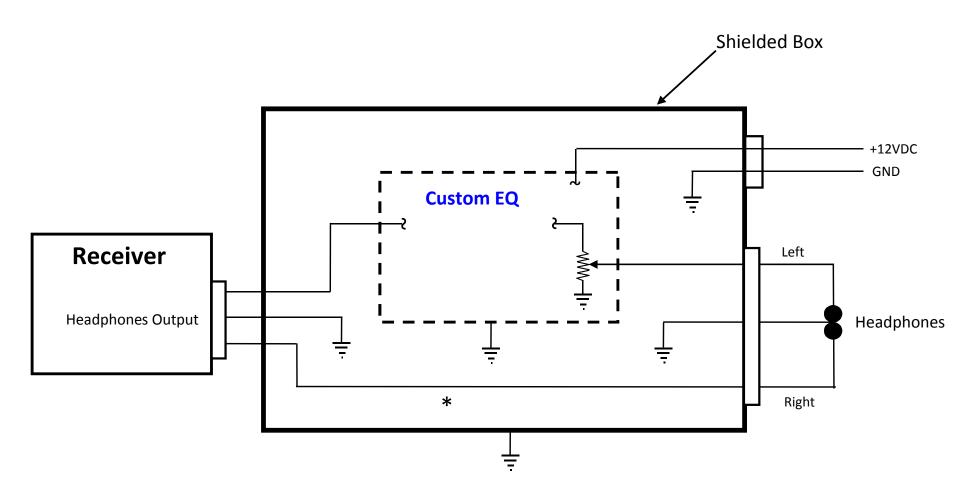
- "Dual Channel Speech Intelligibility Enhancer"
- Dual channel version of MFJ-616
- •\$220 new
- •eHam: no reviews



### **Custom Homebrew Equalizer:**

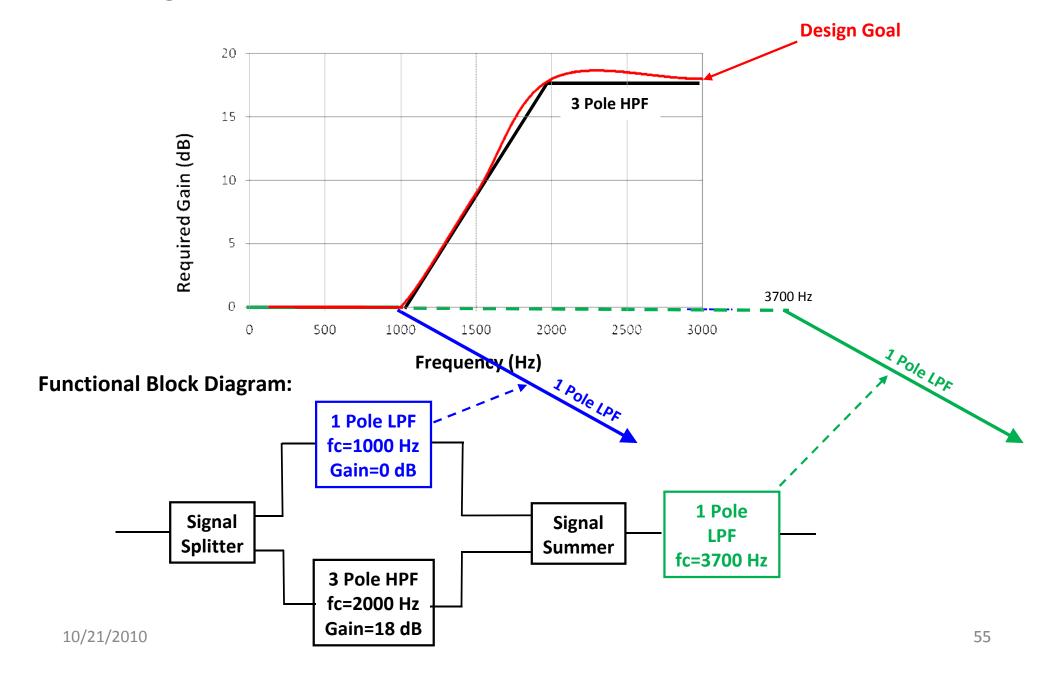
- •Equalizing my left ear response only requires a 2 Band Equalizer
- •Can be made with parts readily available at Radio Shack
  - •2-3 ICs and < 30 R's & C's (<\$50)

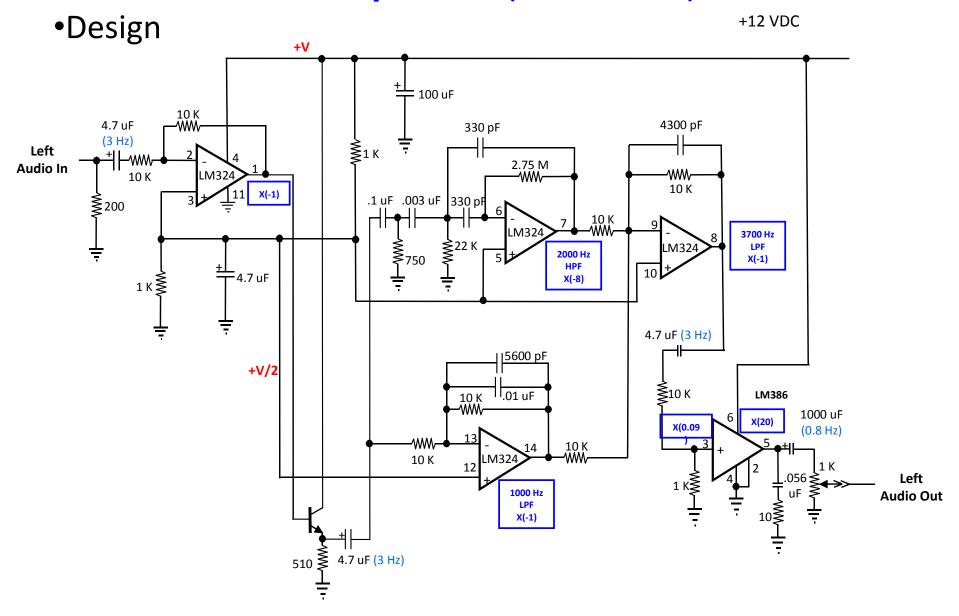




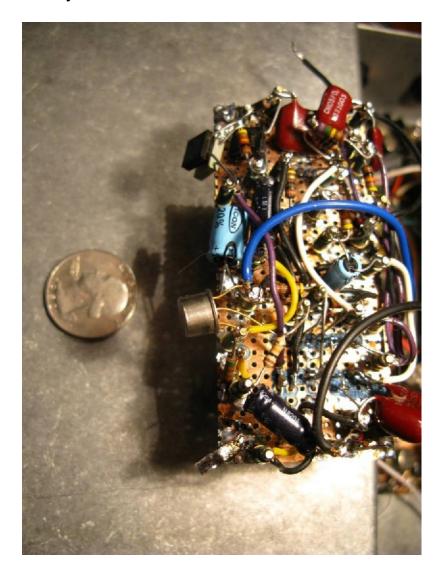
<sup>\*</sup>A second Custom EQ could also be used in the right channel if needed

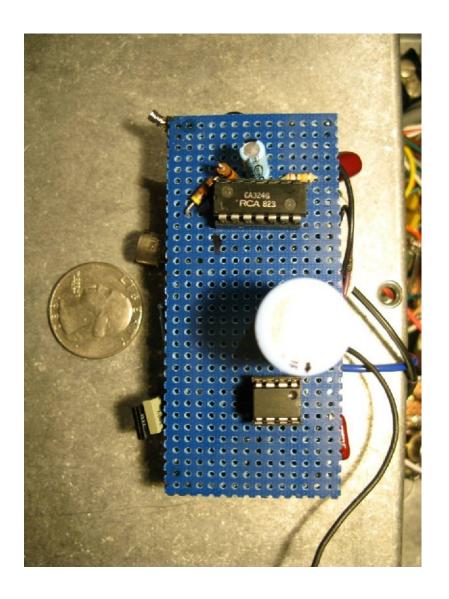
Design



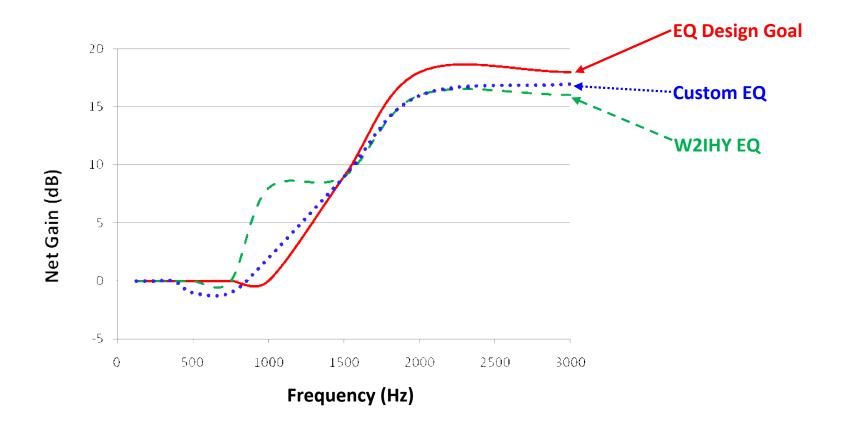


Layout





### **Comparison of results:**



Both Equalizers provide a <u>significant</u> improvement in my ability to copy SSB signals with the left ear!

•Signals with <50% copy =>  $\sim$ 100% copy

### Wrap-up:

- Audio equalizers can be used to improve SSB communications:
  - •Transmit end:
    - Adding "Punch" to the transmitted signal
    - •Compensating for the transmitter operator's speech characteristics
    - •Improving the frequency response of the transmitter hardware
  - •Receive end:
    - Adding "Punch" to the received signal (SNR dependent)
    - •Compensating for the transmitter operator's speech characteristics (SNR dependent)
    - •Compensating for the non-ideal responses of the transmitter and/or receiver hardware (may be SNR dependent)
    - Compensating for the effects of hearing deficiencies on the receive end
- Equalizers will not improve intelligibility of signals with negative SNRs
- Equalizers are becoming standard equipment on new transceivers
   For transmit and receive
- No benefit to using an equalizer for narrowband modes
  CW, PSK31, etc

### Wrap-up (continued):

- •The optimal settings for TRANSMIT applications are <u>very subjective</u> and dependent upon:
  - The application (rag-chew, DX, etc)
  - •The voice characteristics of the transmitter operator
  - The frequency response of the hardware (transmitter & receiver)
  - •The hearing response of the operator on the <u>receive</u> end
  - The best settings are achieved via spectral analysis of the output signal
    - The analysis can be done at the receive end
- Don't need "Studio" quality audio components (20 Hz 20 KHz)
  - •Only need 150 Hz 3 KHz
  - •Big \$\$\$
  - Sometimes they don't sound as good as components designed for "communications" applications