6dB Better than CW

Weak Signal Modes and How They Work

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Why Work with Weak Signals

That rare Dx station
Low power / poor antennas
Cost of big PAs
Extreme propagation - Moonbounce / EME
Traditionally MORSE the weak signal mode of choice.
Compared to SSB voice

So What's wrong with CW / Morse ?

Like having to learn a foreign language
OK when it was compulsory for the Class-A licence, but not now.
Limited range of speeds, ~ 10 - 30WPM.
On-off pulses are difficult in noise
OLD FASHIONED, outdated IMAGE.

What's right with Morse?

The Ear / Brain combination is very good at picking out what it expects to hear Experienced OPs show amazing decoding ability. Rreceiving in the presence of massive QRM and pile ups But not when very weak and buried in noise Half decent source coding Great for contests, pileups, strong QRM Fallback from SSB when things go bad

Some numbers

12WPM morse is ~10 dot/gaps, or symbols, per second, so can be said to occupy 10Hz bandwidth. Noise is proportional to bandwidth. Ear / brain combination filters to 25 – 100Hz for an optimum tone frequency. We're actually wasting capability by sending too slow

Signal / Noise ratio

In any specific signalling bandwidth signal need to be significantly above the noise to absolutely guarantee is it there or not

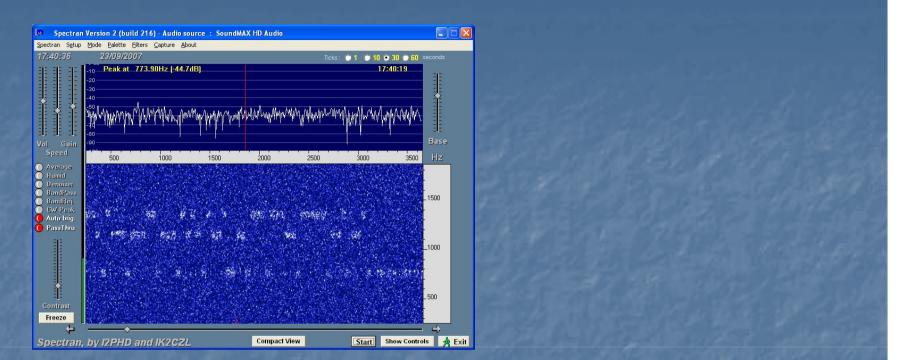
 Human listening needs something like 6dB in the ear bandwidth of 50Hz to do this.

And that is after years of practice !

Ears and software have *about* the same signal detection capability *given optimum settings* and no additional information to the software

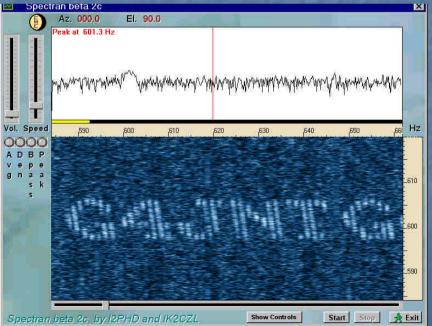
Go Narrow Band

If noise is proportional to bandwidth, why not just go slower? We do ! QRSS at LF uses CW on a visual display in bandwidths down to micro-Hz. Dual / triple Frequency CW But it takes a lot longer to send a message. Minutes or hours for a callsign But allows really weak signal copy.



DFCWi (sending GB3SCX IO80UU)

SMT Hell



Relative Signal/Noise

Speed / bandwidth trade-off, introduces the concept of Normalised S/N referenced to the symbol rate. 24WPM CW in 20Hz bandwidth is identical performance to 1 dot per second in 1Hz QRSS Trade off time-to-send against weak signal detectability. In all cases, we need around 4 – 8dB S/N in that bandwidth

So why not go narrower ?

We haven't got all day. Propagation won't always support it HF Doppler shift and fading, > few Hz VHF / microwave doppler and scatter, 100 Hz Works well at LF. 137kHz generally 0.2 - 0.01Hz is the lower limit. 50s per dot – hours for a callsign.

05 06 07

The first ever QRSS QSO on 73kHz 3 hours for two callsigns 0.04Hz bandwidth

So what can we do ?

Send the information digitally coded onto a Tx waveform

 Optimise signals to work with the lowest (normalised) S/N possible

Data Mode Basics

Chose an optimum Modulation Matched to the RF path Compress the Source Information Reduce the info to be sent to save time Add Error Correction Minimise the chance of wrong decoding Use additional information Time, special message types

Data Modes

Select a modulation type and coding to match the path and wanted data rate Symbol period compatible with propagation Few 10's of Hz for HF If we want a faster data rate, work out a way of stacking multiple slow / narrow carriers increases occupied band.

Modulation Type

Frequency, Phase or Amplitude ? All are valid, but with reservations Phase shift keying, with 0/180 deg is theoretically the best in noise. But practically carrier recovery issues throw away much of the advantage. Amplitude shift keying Works, but is complicated by fading and levels



Frequency Shift is like amplitude shift but with the advantage of comparing one tone against another.

In simplest form two tones. 100% duty cycle, compared with 50% for on-off

IF peak power is the criteria, frequency exchange keying wins over ampl. by 3dB
If TOTAL POWER is the criteria, Amplitude / Frequency shift keying are equal.

Source Compression

Most information is redundant and can be reduced
Callsigns only fit into a few formats
Letters and numbers only
Fitted into 28 bits
Make use of this to reduce the amount of data to be sent

Error Correction

On CW, info is repeated many times But software can do error correction much better than humans repeating things. Add redundant information that is mathematically related to the transmitted data in a special way Allows signals to be 100% copied at relative S/N in the region of 3 - 4dB

Additional Info

Use accurate real time for symbol framing
 Or carrier recovery

Special messages, such as CQ, Roger, and signal reports can be specially coded in very few (strong) bits.

Some Non-weak signal modes

PSK31, 31Hz, 31 Bit/s second – plain text Quite good in pure noise, typically 5dB S/N Lockup time, poor fading and HF Doppler capability, no error correction. `about' the same as CW, overall. RTTY 50B/s – plain text, limited alphabet > 15dB S/N (50Hz). Stop start signalling, very susceptible to noise No error correction

Better Examples

- Modern soundcard modes stack several slow symbol rate tones together to speed up the total net flow
- At the expense of overall occupied bandwidth.
- BUT this is not the same as the noise bandwidth. So avoid the phrase "signal BW"
- 64 tones modulated at 10Hz take up 640Hz, but the noise bandwidth is still 10Hz
- Error correction stacked across tones and time – gives incredible resillience to QRM

WSJT Modes

These are probably the best weak signal performers targetted especially at some of the Dx propagation paths we are interested in.

A lot of thought and design went into optimising the coding and modulation to exactly match practical as well as propagation issues

JT65

Heavily source coded Callsign, Locator and preformatted `QSO messages' Transmited as a sequence of one of 64 tones (6 bits per tone) in 1 minute slots Symbol rate / noise BW 2.7Hz 50% sync. Tone overhead – half the slots Massive error correction capability – 50% lost

JT65 continued

2.7 symbols / second for 48 seconds. Capable of 100% copy in ~4dB S/N Based on the symbol rate. This means decoding at a S/N that is inaudible, and certainly not readable as CW HF to VHF / uWave versions with different tone spacings No great difference in S/N, just signal width.

WSPR

Preformatted message with callsign, locator and Tx power – for personal beacons
4 tones, 1.5Hz spacing and symbol rate
50% sync overhead,
4dB S/N in that bandwidth for perfect copy

Opera

On/Off keying, so an inherent 3dB penalty on peak power rating over FSK
Many speed / data rate settings
Heavy source coding and error correction
Only slightly worse normalised S/N perfomance as WSJT modes based on mean power.

But... 3dB worse if based on <u>peak</u> power.

Summary so far...

 Optimise data sent, add error correction, match modulation + speed the path.
 Normalising data rate to the bandwidth, we can show that a properly designed data mode can show about 6dB better than CW with a highly experienced operator.

Others, Keyboard Modes Olivia, MFSK63, ROS, and many, many more. Need to be faster Often stacked paralled carriers for keyboard typing speed Can be quite wide (SSB bandwidth) Inherent delay for FEC But most are VERY robust, allowing copy on almost inaudible tones.

Compare CW effective Data rate

Assume 18WPM in 30Hz ear/brain bandwidth (a 'good' operator) Needs about 10dB S/N in this bandwidth. A word has 5 chars, 5 (ish)bits / char (plain text) so about 7.5Bits /second equiv data rate Repeating the message for redundancy gives around 3 Bits/second overall. Now normalise to reference bandwidth • 3 Bits/second in 30Hz = 0.1B/s/Hz.

JT65 Data Rate

Pre-format two callsigns plus locator Before compression (if it were to be sent in CW it would be plain text) needs 80 bits Sent in a one minute window = 1.33 Bits/s Normalise to the Noise B/W of 2.7Hz Means 0.5 Bits/second/Hz Which is 5 times better than CW can manage, or 7dB. Calling it 6dB looks more reasonable!

A few caveats

Most data modes do not allow arbitrary trade off of speed vs. Bandwidth

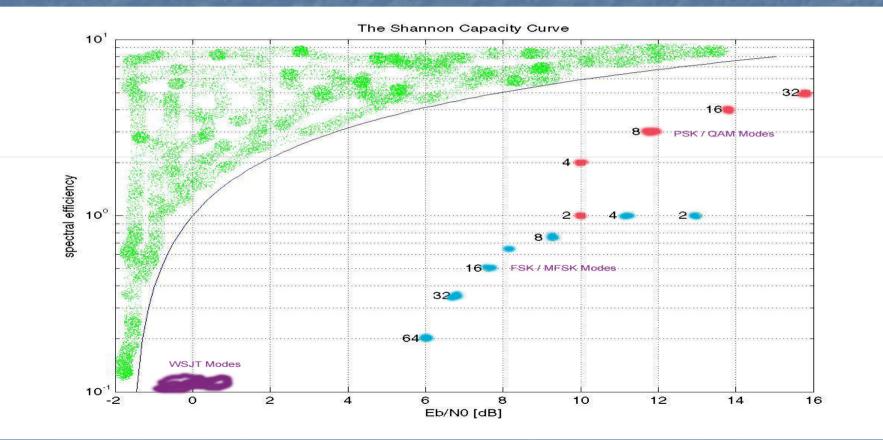
Telling the other end what your are using is not easy.

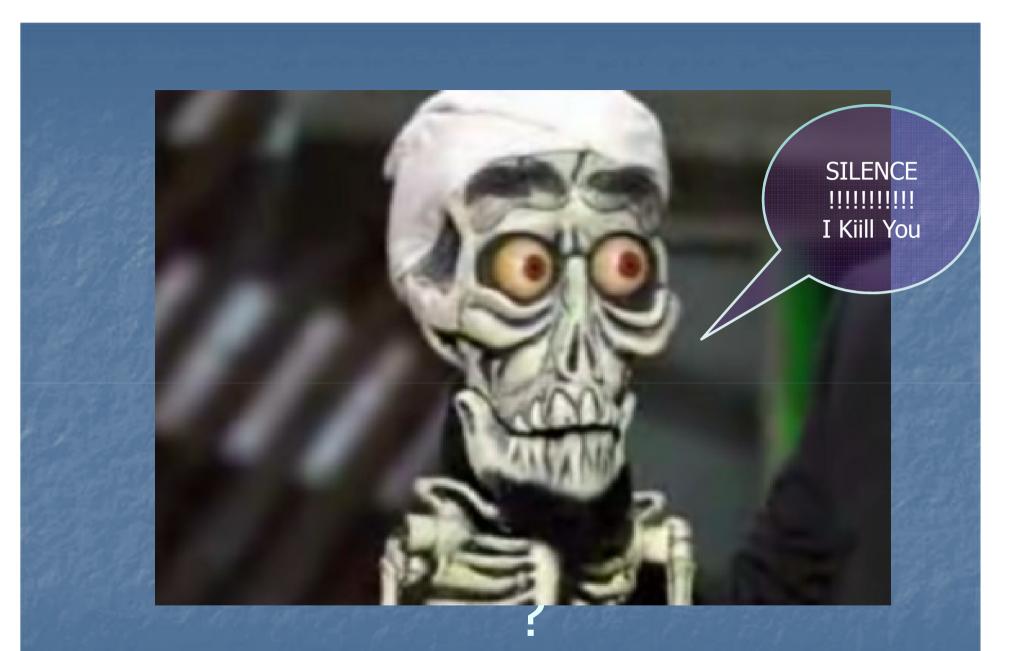
So like-for-like comparisons aren't easy
Limits on Occupied bandwidth

3kHz SSB radios, regulatory, licences etc.

But shows what can be done if the software can be written

And Finally - what could be possible





http://www.g4jnt.com/SNDemo.exe

