Applying Codec2 To Amateur Satellite Operations

Bruce Robertson, VE9QRP Amsat Symposium, 2012

QSO with ZL3IN in Early 2011



About 45 min. conversation



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The complete recording of both sides of the conversation could fit on a 1.4MB floppy disk



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- Attuned to the human voice



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- Bruce Perens, K6BP, noted the absence of a free and open-source low-bitrate voice codec
- Encouraged David Rowe, VK5DGR, to repurpose the work in his Ph.D. thesis to this end
 - Built community and code base
 - Won ARRL technical innovation award in 2012



Free and open source



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- Codecs from 3200 bit/s to 1200

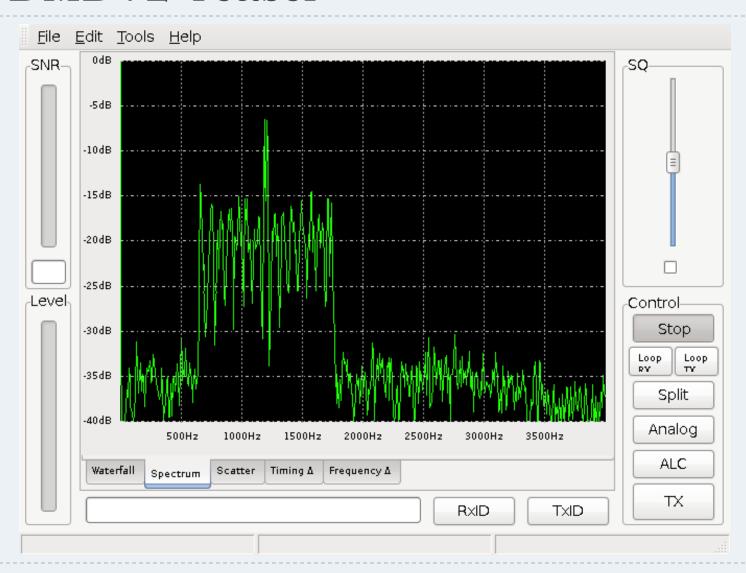
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- Modems:
 - ▶ 1400 bit/s FDMDV
 - ▶ 4800 and 2400 (early) GMSK



FDMDV2 Teaser



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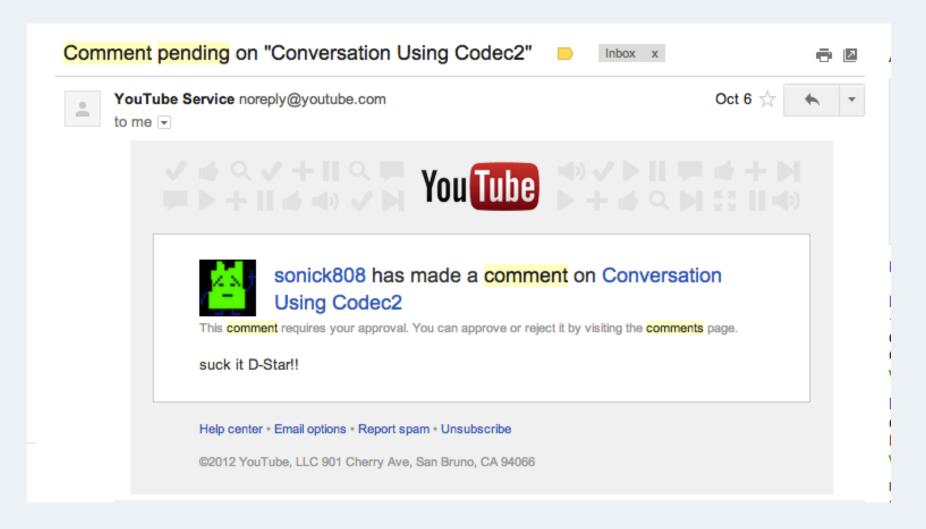


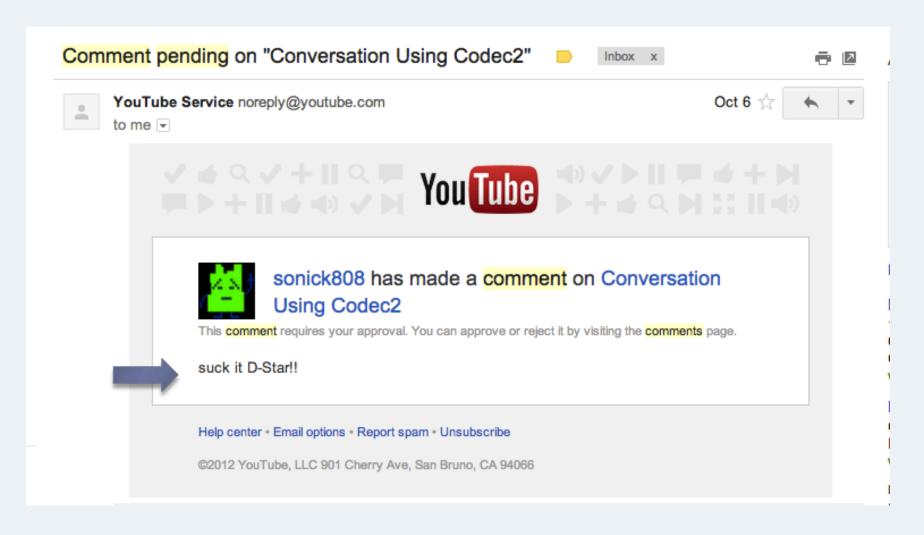
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- But many satellite stations have a computer attached







- ▶ Not about
 - Hijacking D-Star



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 - Hijacking D-Star
- Current emphasis on HF voice



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Kristoff Bonne, ONIARF



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For example?



▶ FDMDV through FM birds



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- ▶ FDMDV through FM birds
- TWO FDMDV channels (x 1050 Hz) through one FM bird
 - ▶ Who's going to call that an Easy-Sat? ☺
- FDMDV signal acquisition and doppler correction



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- Codec2 can bridge the voice / digital gulf
 - Allows a functionally digital mission
 - ▶ To assist in improving voice communications
 - ▶ To provide voice communications
 - No more 'beep boxes'
- Digital comms. should be arranged with this possibility in mind
 - ▶ 1200 AFSK won't cut it
 - 2000-2400 BPSK/QPSK?



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- Offers interesting research problems for undergraduate cubesat programs:
 - E.g., what sort of modulation technique and how much error correction would be helpful in LEO?



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- Challenge:
 - Including efficient codec2 modulation code in on-board computing system



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 - ► (Twit's live stream to shoutcast)
 - gst-launch souphttpsrc location=http://twit.am:80/listen !
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- Introduce random errors into FEC-encoded Codec2 to evaluate interleaving approaches
 - Extra points: make a website that crowd-sources the scoring of various approaches



Last Word to the Codec

- ▶ 2.1 kB message at 1200 bits/s
- Come see me for a record/playback session with my laptop

