

Deep Redundancy for the Opus Codec

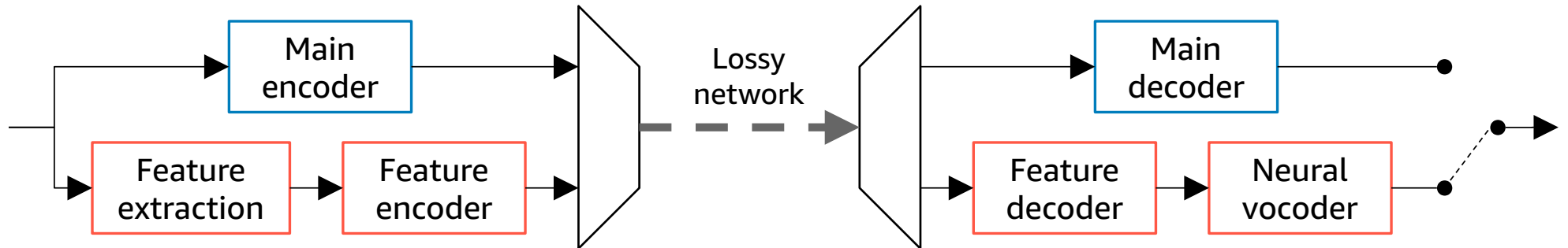
draft-valin-opus-dred-05

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IETF 119

DRED Recap

- Code large amounts of redundant audio in Opus packet
 - Use DNN to maximize compression
 - Can code 1 second per 20-ms packet (50x)



Changes Since IETF 118

- Added extended offset
 - Optional extra byte signals up to 20 second offset
 - Used to trim silence (more efficient, better DTX)
 - Potential use in SFU just after switch
- Added Qmax field
 - Makes it possible to cap the quantizer
 - Variable size
 - Costs one bit when unused

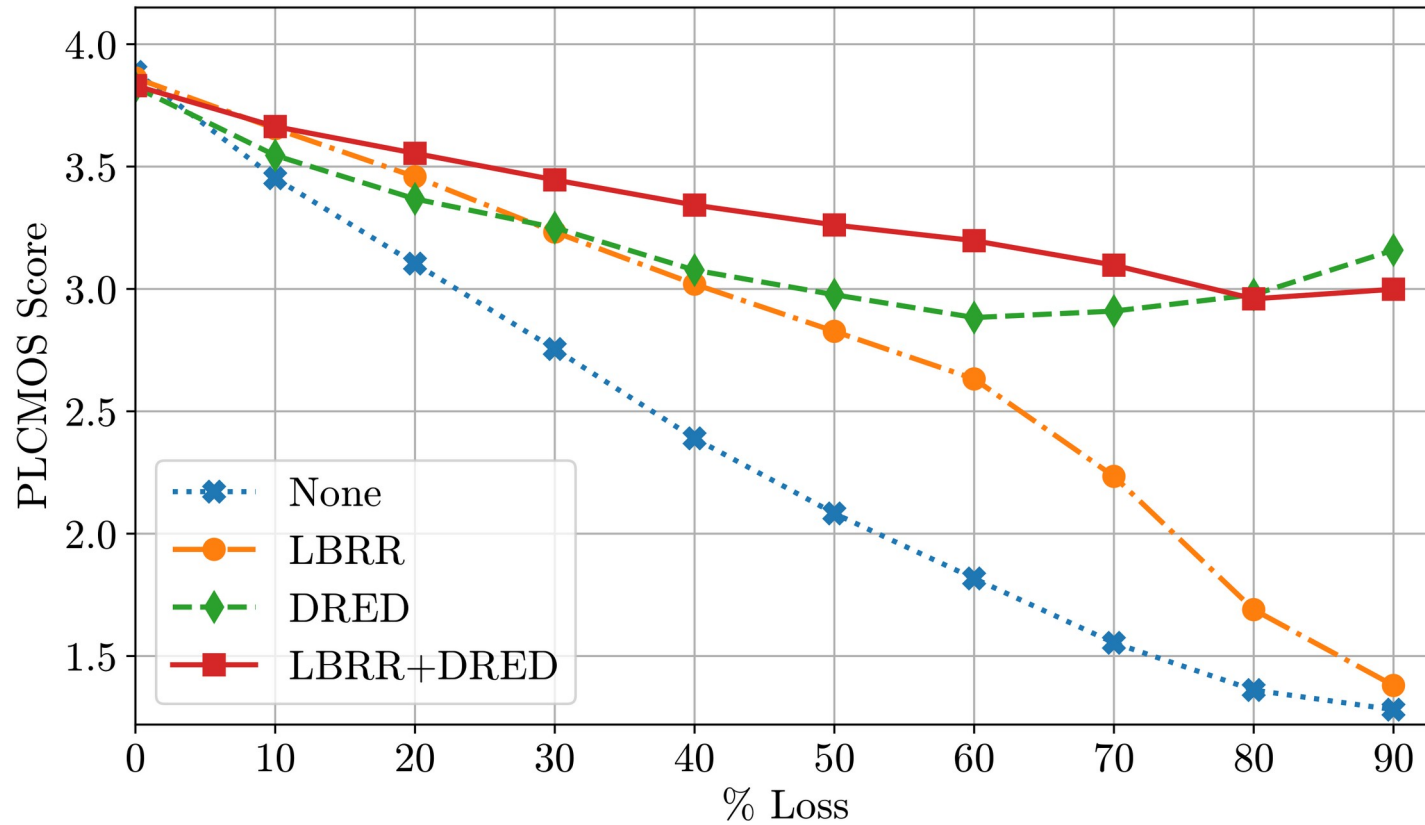
Normative Aspects

- Normative spec for bits-to-feature decoder
 - All decoder weights are frozen
 - Publish as simple binary format
 - Definition of the acoustic features
 - How do we specify neural pitch estimator?
- Encoder is left unspecified
- Minimal constraints on vocoder

Implementation Update

- Changes landed in new Opus main branch
- Released latest implementation in Opus 1.5
 - <https://www.opus-codec.org/demo/opus-1.5/>
 - DRED disabled by default
- WebRTC patch set with DRED support
 - <https://github.com/xiph/webrtc-opus-ng/>

Results



Open Questions

- Should there be a maximum duration allowed?
 - Technically we could do up to ~10 minutes
 - Proposal: no hard limit, since receiver can ignore the rest
- What are the lowest and highest useful bitrates?
 - Currently support 10 to 100 kb/s for 1 second redundancy
 - Equivalent to 200 b/s to 2 kb/s effective bitrate