

row-mp3: An Enhanced MP3-Compatible Audio Codec

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Outline

- ▶ Introduction and Motivation
- ▶ Coding Error
- ▶ Analysis
- ▶ Synthesis
- ▶ Example: “row-mp3”

Introduction and Motivation

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Introduction and Motivation

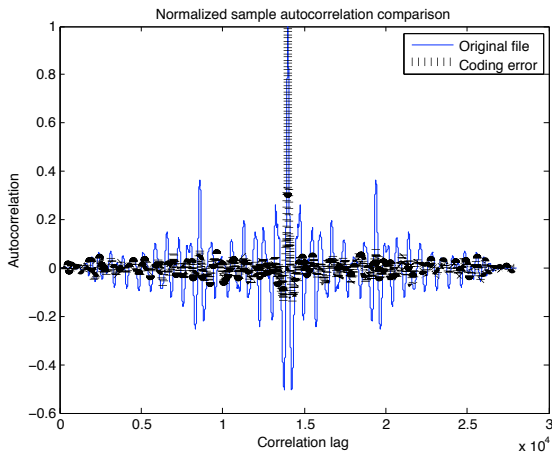
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- ▶ Technique: Find a way to represent the coding error with a small amount of data and include the information in the coded audio file
- ▶ Proposal: Store frame-by-frame, per-critical-band residual levels in the audio codec's metadata and re-synthesize the coding error as colored noise when decoding

Coding Error

- ▶ Achieving lower data rates requires some information loss
- ▶ We can define coding error as $(original\ audio) - (coded\ audio)$
- ▶ Tends to be noisy \triangleright
- ▶ Modeling as colored noise is cheap



Residual Analysis: Spectral Flux

- ▶ Idea: Model only the non-stationary component of the error
- ▶ Simple method: Spectral flux, defined as

$$SF(n) = \sqrt{\sum_{k=0}^{N-1} (|X[n, k]| - |X[n-1, k]|)^2}$$

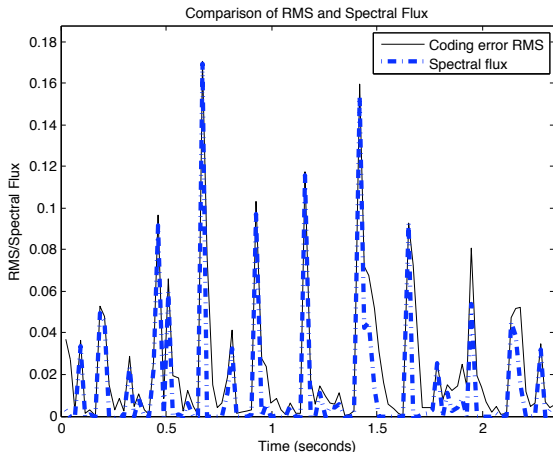
- ▶ Stationary signal components get subtracted out
- ▶ Roughly speaking,

$$SF(n) \propto \text{RMS}(x[n])$$

- ▶ Full proof is in the paper
- ▶ Proportionality only holds for Gaussian noise and non-overlapping rectangular windows

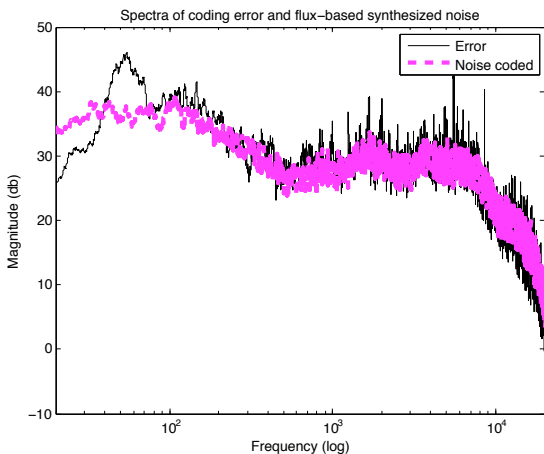
Residual Analysis: Spectral Flux

- ▶ Coding error does not satisfy proportionality criterion
- ▶ The proportionality still roughly holds in practice



Residual Analysis: Spectral Flux

- ▶ To determine coloring, evaluate the flux on a per-band basis
- ▶ Band levels tended to change too rapidly from frame-to-frame
- ▶ However, RMS proportionality holds in practice and makes this technique useful ▷



Residual Analysis: Smoothed Cepstrum

- ▶ Obtain spectral envelope by windowing the real cepstrum and taking the DFT

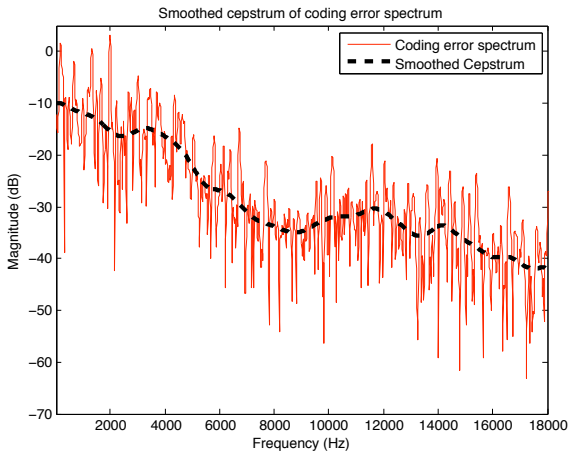
$$C[n] = \Re \left(\frac{1}{N} \sum_{k=0}^{N-1} \log(|X(k)|) e^{j2\pi nk/N} \right)$$

$$E[k] = \Re \left(\sum_{n=0}^{N-1} w[n] C[n] e^{-j2\pi nk/N} \right)$$

- ▶ Works well for relatively peak-free spectra
- ▶ Per-band level can be found by averaging over bins in band

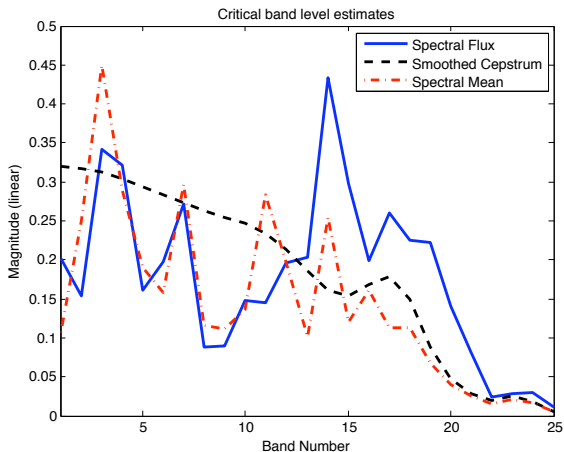
Residual Analysis: Smoothed Cepstrum

- ▶ Generally results in band levels which are “smooth” from band to band and frame to frame ▷



Residual Analysis: Comparison

- ▶ Flux is analytically “clean”, but varies rapidly because it is intentionally uncorrelated
- ▶ Smoothed cepstrum provides a reasonable estimate which is smoother in time and band



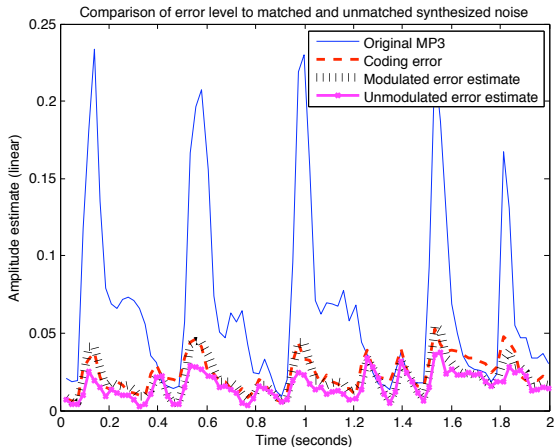
Residual Synthesis

- ▶ Generate coding error representation by applying critical band envelopes to a random spectra ▷
- ▶ Envelope differences from frame-to-frame cause coloring discontinuities
 - ▶ We can generate any amount of colored noise by generating a larger spectrum
 - ▶ So, create additional noise per-frame and crossfade
- ▶ Transients in the residual result in frames of noise in the error representation
 - ▶ Traditional methods for detecting and representing transients are not effective
 - ▶ The coded audio and coding error's envelopes are similar
 - ▶ We can modulate residual representation with the coded audio's envelope

Residual Synthesis

- ▶ We can parametrize the amount of envelope modulation by ▶

$$y[n] = ((1 - \alpha) + \alpha L[n]) x[n]$$



Implementation: “row-mp3”

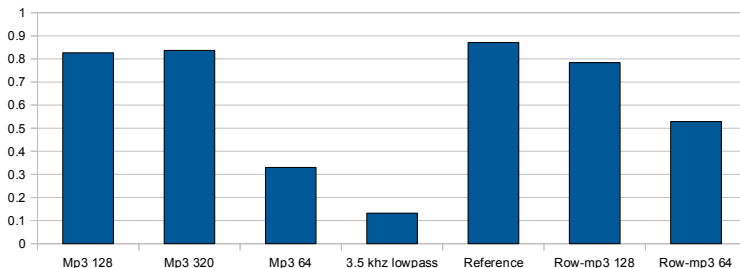
- ▶ The MP3 codec is highly pervasive but somewhat out-of-date
- ▶ To allow backwards-compatibility, we can store information in the ID3 (metadata) tag
- ▶ “row-mp3”-aware decoders can use the information, while others will simply ignore it
- ▶ Including per-frame critical band levels results in a relatively small data overhead
 - ▶ For example, with a 23.2 ms frame size and 8-bit quantized band level values we have

$$(.0232) * (8) * (25) = 8.6 \text{ kbit/s/channel}$$

- ▶ Data overhead can be reduced by using different quantization schemes or compression such as Huffman coding

Implementation: “row-mp3”

- ▶ Created a simple MUSHRA-like web-based test to determine codec's effectiveness
- ▶ row-mp3 files used spectral flux method with no envelope modulation
- ▶ 60 subjects tended to rate the row-mp3 version about 150% better for low MP3 bit rates
- ▶ Further, more controlled testing with all error analysis and synthesis methods is needed



Conclusions

- ▶ Audio coding error can be effectively modeled as colored noise
- ▶ Flux provided a “theoretically-sound” coloring estimate
- ▶ Cepstral smoothing works better in practice
- ▶ Synthesis by scaling random spectra
- ▶ Cross-fading and interpolation prevented coloring discontinuities
- ▶ “Level-modulated” error estimate helped prevent smeared transients
- ▶ row-mp3 codec and accompanying listening tests suggest feasibility

Future work

- ▶ Investigating the optimal number and spacing of bands
- ▶ Testing the effectiveness of other analysis techniques
- ▶ Evaluating different methods for dealing with transients
- ▶ Applying similar techniques to spectral modeling and other processes with residual
- ▶ Implementing inclusion schemes in other audio codecs
- ▶ Generating residual levels solely from the coded audio (as a sound enhancement)

Acknowledgements

- ▶ Jieun **O**h and Isaac **W**ang for creating the “row-mp3” codec
- ▶ Prof. Marina Bosi for her instruction in the field of audio coding
- ▶ Prof. Julius Smith for helpful advice and discussion on various topics

Sound examples and code

`http://ccrma.stanford.edu/~craffel/software/noise/`

`http://ccrma.stanford.edu/~craffel/software/rowmp3/`