

SPECTRAL DELAY FILTERS WITH FEEDBACK AND TIME-VARYING COEFFICIENTS

DAFx-09 PRESENTATION

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CASCADED FIRST-ORDER ALLPASS FILTERS

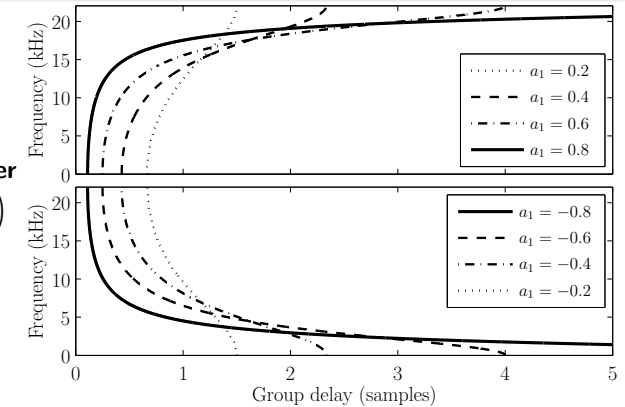
Frequency-Dependent Delay Produced by a Filter

- Described by the group delay
- Cascading of filters sums the individual group delays!

Positive coefficient

First-order allpass filter
 $\left(\frac{a_1 + z^{-1}}{1 + a_1 z^{-1}} \right)$

Negative coefficient



INTRODUCTION

Spectral Delay Filtering?

- Frequency-dependent delay **Demo: Original**
- Earlier with STFT **Filtered**
 - D. Kim-Boyle, "Spectral Delays with Frequency Domain Processing," in *Proc. DAFx-04*, pp. 42–45, Naples, Italy, Oct. 2004.
 - X. Amatriain, *An Object Oriented Metamodel for Digital Signal Processing with a Focus on Audio and Music*, PhD. thesis, Universitat Pompeu Fabra, Barcelona, Spain, 2004.

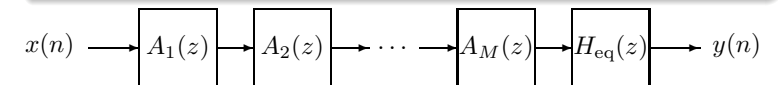
Alternative Approach to STFT?

- Inherently better resolution & large delays with less memory?
⇒ Solution: Cascaded allpass filters
 - V. Välimäki, J. S. Abel and J. O. Smith, "Spectral Delay Filters," *J. Audio Eng. Soc.*, vol. 57, no. 7/8, pp. 521–531, July/Aug. 2009.

CASCADED FIRST-ORDER ALLPASS FILTERS

Spectral Delay Filter (SDF)

- Cascade of several first-order allpass filters ($A_i(z)$)
- Includes also an optional equalization filter ($H_{eq}(z)$)

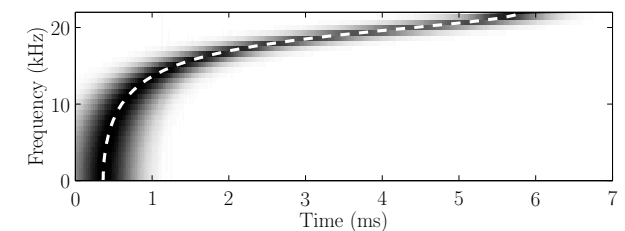


Impulse response, $a_1 = -0.9$, no EQ:

$M = 256$

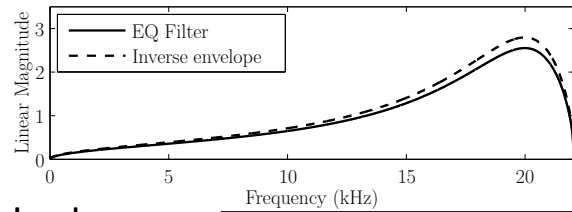
$M = 512$

$M = 1024$



(plot with $M = 64$ and $a_1 = 0.6$)

EQUALIZATION FILTER DESIGN



Design Example:

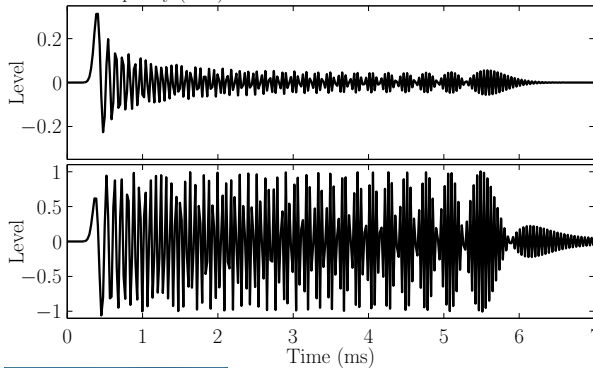
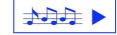
- $M = 1$
- $a_1 = 0.6$
- Without scaling

Impulse response,
 $M = 1024,$
 $a_1 = -0.9:$

Without EQ

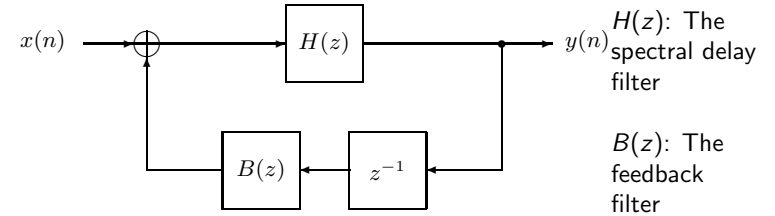


With EQ



(plots with $M = 64$ and $a_1 = 0.6$)

SPECTRAL DELAY FILTERS WITH FEEDBACK



$H(z)$: The spectral delay filter

$B(z)$: The feedback filter

Stability Conditions

- Both filters causal and stable
- The loop gain less than unity at all frequencies

Feedback Filter Design Issues

- The gain of $H(z)$ determined solely by the EQ filter
- ⇒ Sufficient attenuation must be applied by the feedback filter
- ⇒ Unsophisticated (low-order) filter designs often impractical

STRETCHED SPECTRAL DELAY FILTERS

Multirate Version of the Basic SDF

- Every unit delay replaced with K unit delays (in EQ filter, too!)

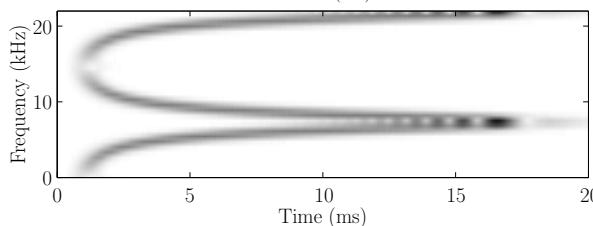
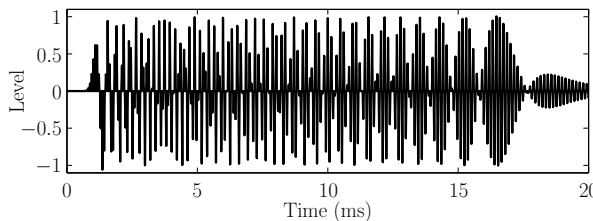
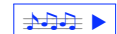
Impulse response,
 $M = 1024,$
 $K = 3,$
 $a_1 = -0.9:$



"Alien hiccup"
Original



Filtered



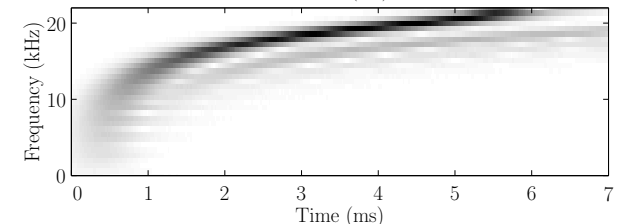
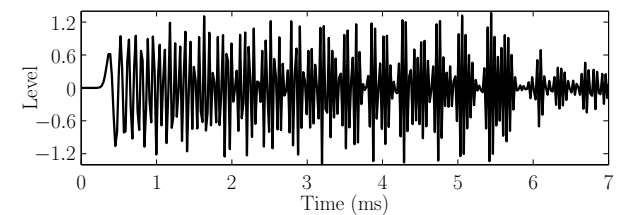
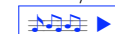
(plots with $M = 64$ and $a_1 = 0.6$)

SPECTRAL DELAY FILTERS WITH FEEDBACK

Design Example

- Feedback filter $B(z) = c(1 + z^{-1})$

Impulse response,
 $M = 256,$
 $K = 1,$
 $a_1 = -0.9,$
 $c = 1/436$



(plots with $M = 64,$ $a_1 = 0.6,$ $c = 1/23$)

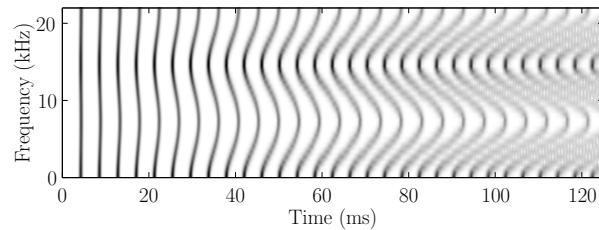
TIME-VARYING SPECTRAL DELAY FILTERS

Time-Varying Spectral Delay Filtering?

- Coefficients vary in time
- Lively, more dynamic effects obtained

Stability Conditions

- The coefficients in the range $[-1, 1]$, inclusively, for all $K!$
- Same condition for a time-varying EQ filter
- Gain depends on the modulation (and the input signal)



CONCLUSIONS



Summary

- Frequency-dependent delay implemented with allpass filters
 - Inherently good resolution
 - Large delays obtainable with a low-order filter
- EQ filter to emphasize the slow (soft) part of the chirp
- Multirate structures \Rightarrow Simultaneous chirps
- Feedback structures \Rightarrow Series of chirps
- Time-varying structures \Rightarrow Lively, dynamic effects

Further Pointers

- The conference paper
 - Background paper: V. Välimäki, J. S. Abel and J. O. Smith, "Spectral Delay Filters," *J. Audio Eng. Soc.*, vol. 57, no. 7/8, pp. 521–531, July/Aug. 2009.
- Companion page: <http://www.acoustics.hut.fi/publications/papers/dafx09-sdf/>

TIME-VARYING SPECTRAL DELAY FILTERS

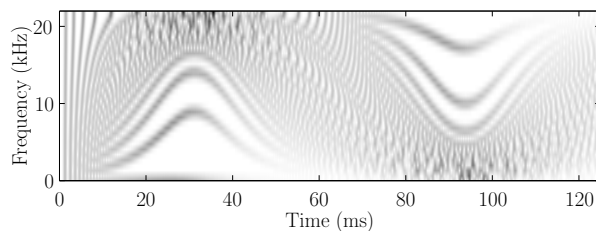
Example: $K = 1$, with coefficients modulated with an 8 Hz sine having amplitude 0.9, without EQ filter, and with a feedback path with a constant multiplier of $B(z) = 0.99$

$M = 256$ $M = 512$ $M = 1024$

Impulse response Original

Drumming pattern

Plot: Impulse response with $M = 64$



Paper on Audio-Rate Modulated SDF

- J. Kleimola *et al.*, "Sound Synthesis Using an Allpass Filter Chain with Audio-Rate Coefficient Modulation," in *Proc. DAFX-09*...

