



Virtual Analog Modeling Research at Aalto University

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Edinburgh, UK

Feb. 27, 2013

Aalto University

- Formed in 2010 as a merger of 3 universities, including the Helsinki University of Technology (HUT/TKK)
- Otaniemi campus located in the city of Espoo
 - About 10 km from the Helsinki city center



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Otaniemi Campus, Espoo, Finland



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School of Electrical
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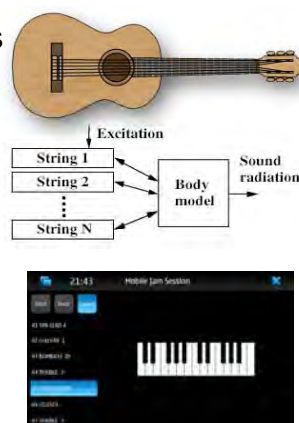
Audio Signal Processing Research Group

- Professor: Dr. Vesa Välimäki
- Senior researchers:
 - Dr. Henri Penttinen
 - Dr. Ole Kirkeby
- Postdoc researchers:
 - Dr. Heidi-Maria Lehtonen
 - Dr. Rémi Mignot (IRCAM)
- 6 researchers (PhD students):
 - S. D'Angelo, S. Oksanen, R. de Paiva, J. Parker, J. Rämö, H. Tuominen
- Visitors
 - From Italy, Estonia, Denmark, ...
- Main funding sources
 - Academy of Finland
 - EU
 - GETA, CIMO
 - Companies (Nokia, Sandvik)



Current Research Topics

- Physical modeling of sound sources
 - Musical instruments and noise sources
- Modeling of analog music technology
 - 'Virtual analog' models
- Sound synthesis and effects processing algorithms
- Headphone audio
- Digital filters for audio processing

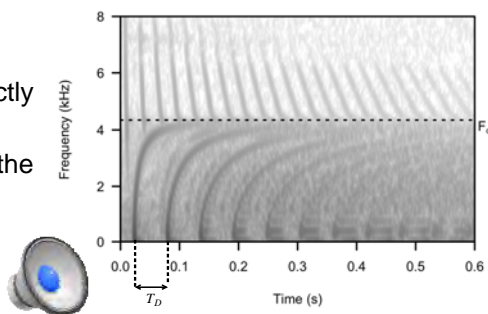


Spring Reverberation



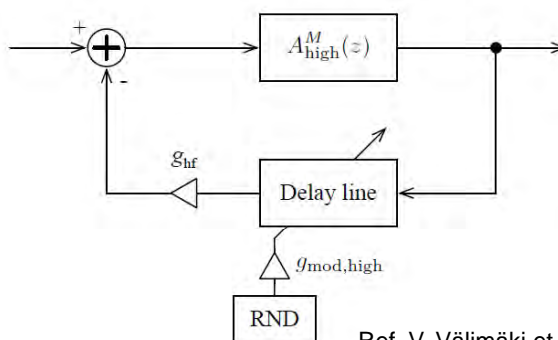
- Spring reverberators are an early form of artificial reverberation

- Reminiscent of room reverberation, but with distinctly different qualities
- Our team has characterized the special sound of the spring reverberator, and modeled it digitally



Parametric Spring Reverberation Model

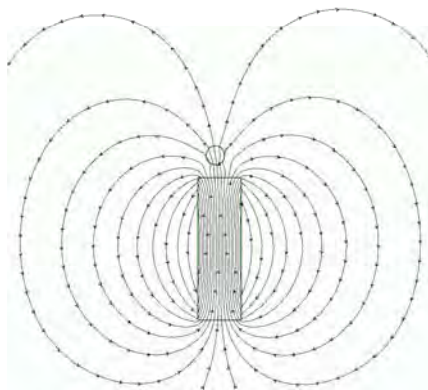
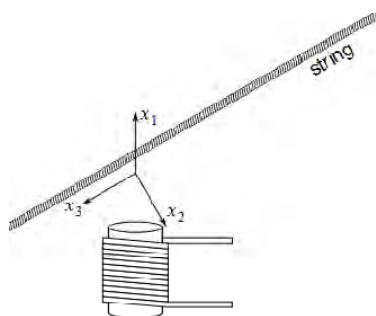
- Many (e.g. 100) allpass filters produce a chirp-like response
- A feedback delay loop produces a sequence of chirps
- Random modulation of delay-line length introduces smearing



Ref. V. Välimäki et al., JAES, 2010.

Magnetic Induction in Guitar Pickup

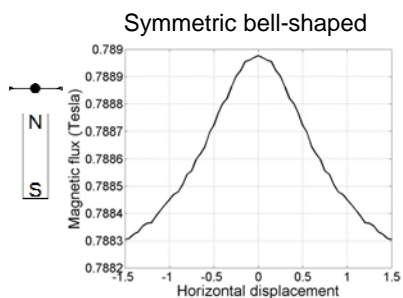
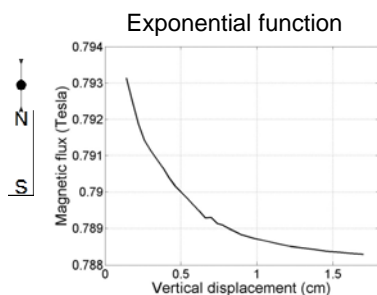
- String proximity increases the magnetic flux
- The change causes an alternating current in the winding



Ref. Paiva et al., JAES, 2012.

Pickup Nonlinearity

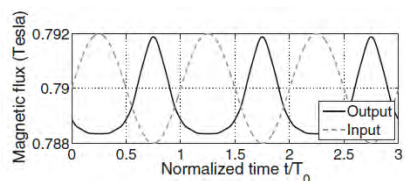
- Sensitivity is different for the vertical and horizontal polarizations
- 2-D FEM simulations using Vizimag



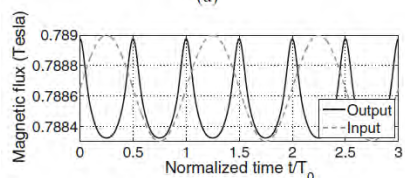
Ref. Paiva et al., JAES, 2012.

Pickup Nonlinearity

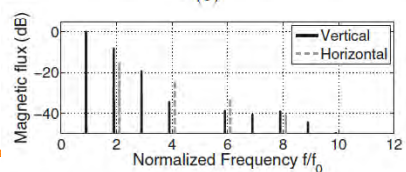
- a) String displacement in the **vertical** direction leads to harmonic asymmetric distortion (*all* harmonics)
- b) String displacement in the **horizontal** direction leads to harmonic symmetric distortion (*even* harmonics)



(a)



(b)

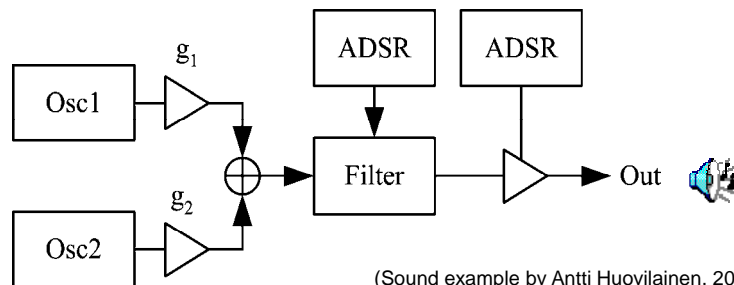


(c)

Ref. Paiva et al., JAES, 2012.

Digital Subtractive Synthesis

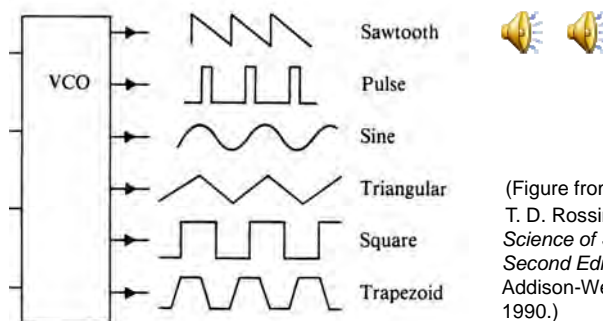
- Emulation of analog synthesizers of the 1970s
- One or more oscillators, e.g., an octave apart or detuned
- Second- or fourth-order resonant lowpass filter
- At least two envelope generators (ADSR)



(Sound example by Antti Huovilainen, 2005)

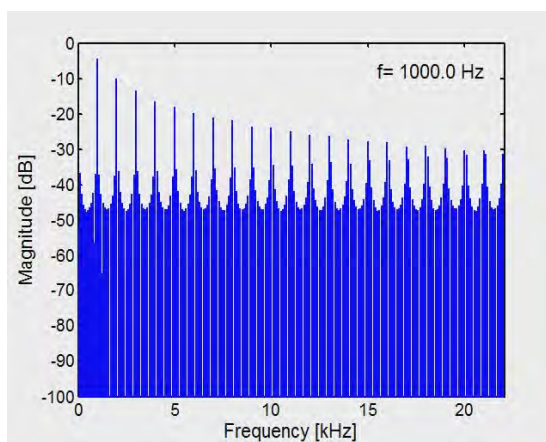
Oscillators in Subtractive Synthesis

- Usually periodic waveforms
 - All harmonics or only odd harmonics of the fundamental
- Digital implementation must suppress aliasing



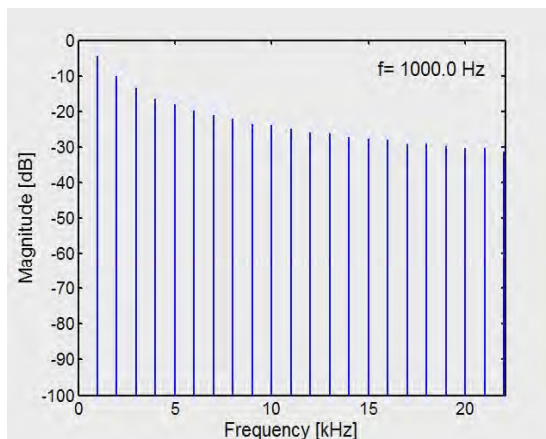
Aliasing – The Movie

- Trivial sampling of the sawtooth signal
- Harsh aliasing particularly at high fund. frequencies
 - Inharmonicity
 - Beating
 - Heterodyning



No Aliasing

- Additive synthesis of the sawtooth signal
- Contains harmonics below the Nyquist limit only



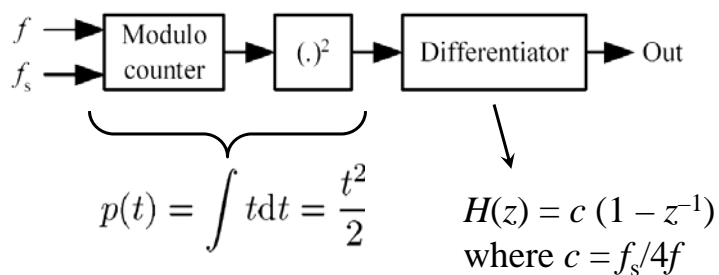
Video by Andreas Franck,
2012

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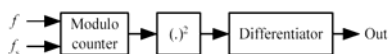
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Differentiated Parabolic Wave Algorithm

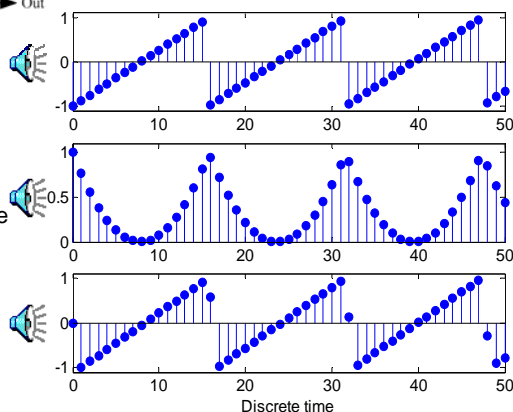
- A method to produce a sawtooth wave with reduced aliasing (Välimäki, 2005)
 - 2 parameters: fundamental frequency f and sampling frequency f_s



Signal Generation in DPW Algorithm

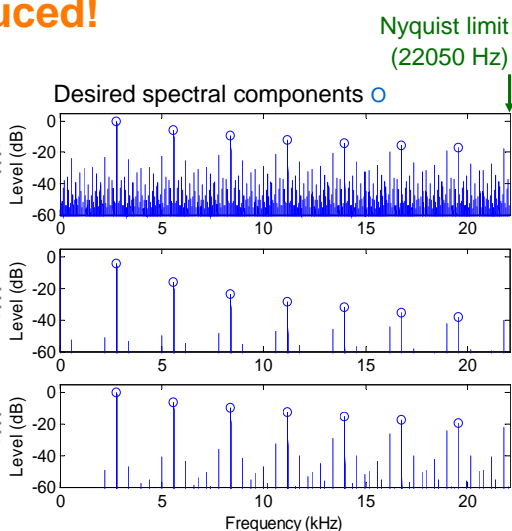


- Output of modulo counter $x(n)$
 - A 'trivial' sawtooth wave
- Squared signal $x^2(n)$
 - Piecewise parabolic wave
- Differentiated signal $c [x^2(n) - x^2(n-1)]$
 - Difference of neighbors



Aliasing is Reduced!

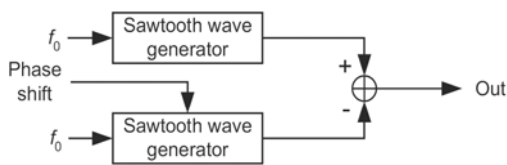
- Spectrum of modulo counter signal $x(n)$
- Spectrum of squared signal $x^2(n)$
- Spectrum of differentiated signal $c [x^2(n) - x^2(n-1)]$



Rectangular Pulse Generation Using DPW

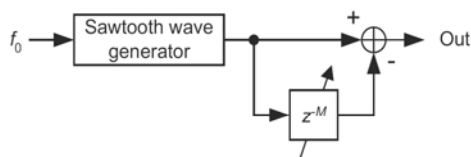
- Two alternative methods (Välimäki & Huovilainen 2006)

(a) Subtract two sawtooths



(a)

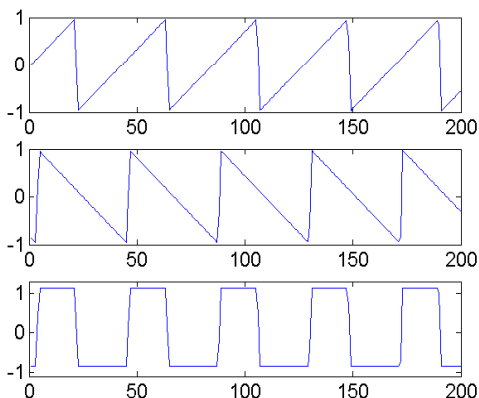
(b) Use an FIR comb filter to generate the phase shift, then subtract



(b)

Rectangular Pulse Generation Using DPW

- Sawtooth #1
- Sawtooth #2:
Delayed and inverted
- Rectangular pulse



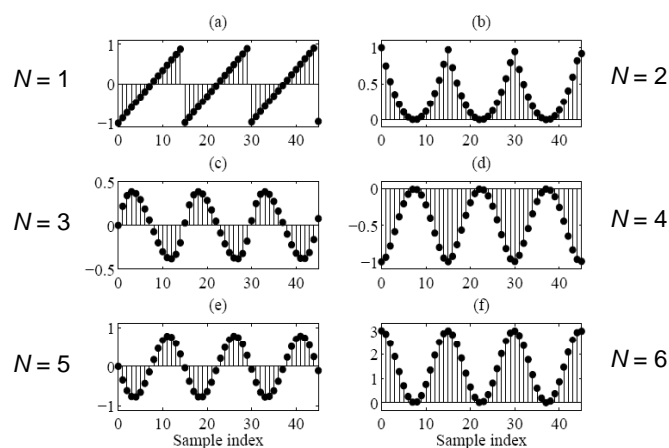
Higher-order DPW Oscillators

- Trivial sawtooth can be integrated multiple times (Välimäki *et al.*, 2010)

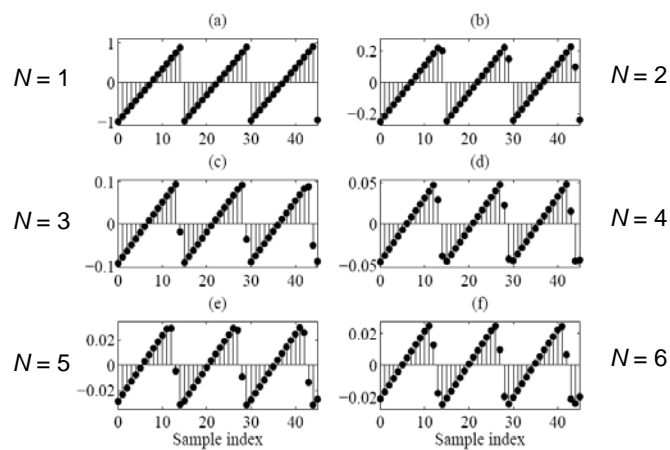
Polynomial order	Polynomial function
$N = 1$	x
$N = 2$	x^2
$N = 3$	$x^3 - x$
$N = 4$	$x^4 - 2x^2$
$N = 5$	$x^5 - 10x^3/3 + 7x/3$
$N = 6$	$x^6 - 5x^4 + 7x^2$

The polynomial signal must be differenced $N - 1$ times and scaled to get the sawtooth wave

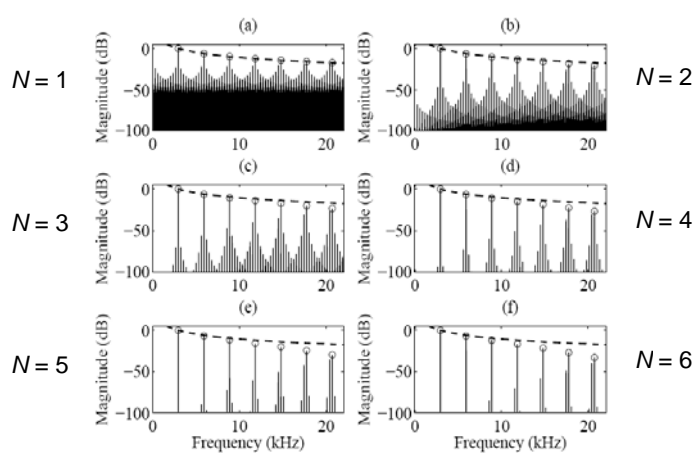
Integrated Polynomial Waveforms



Differenced Polynomial Waveforms



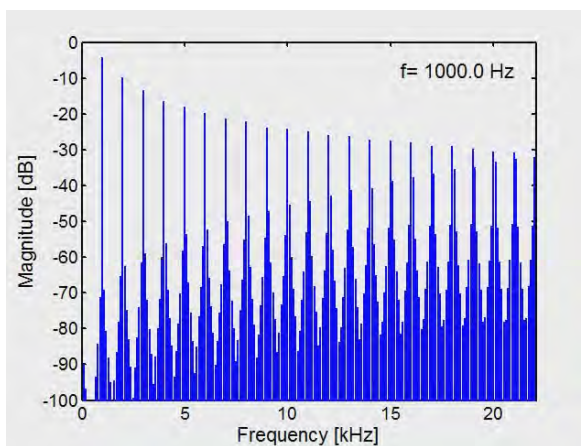
Spectra of Differenced Waveforms





DPW Sawtooth Sweep

- One integration and derivation



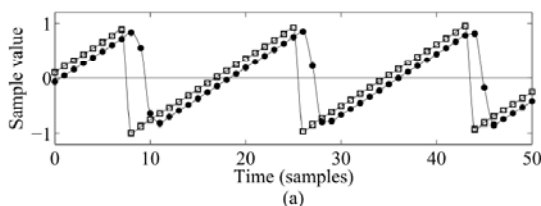
Video by Andreas Franck,
2012

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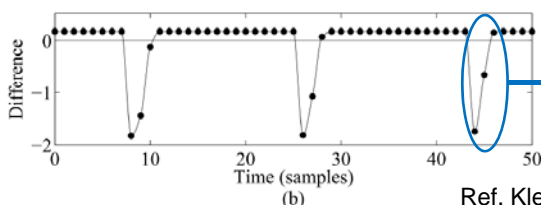
Polynomial Transition Region (PTR)

- The PTR algorithm implements DPW efficiently and extends it



□ Trivial sawtooth
(modulo counter)

• DPW waveform



← Constant offset

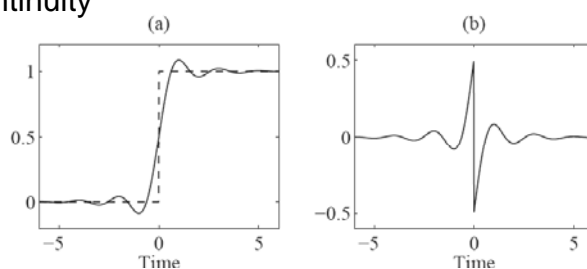
← Sampled polynomial
transition region

Ref. Kleimola and Välimäki, 2012.



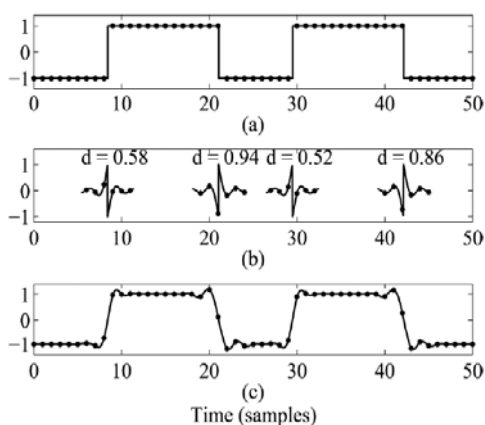
BLEP Method

- BLEP = Bandlimited step function (Brandt, ICMC'01), which is obtained by integrating a sinc function
 - Must be oversampled and stored in a table
- BLEP residual samples are added around every discontinuity



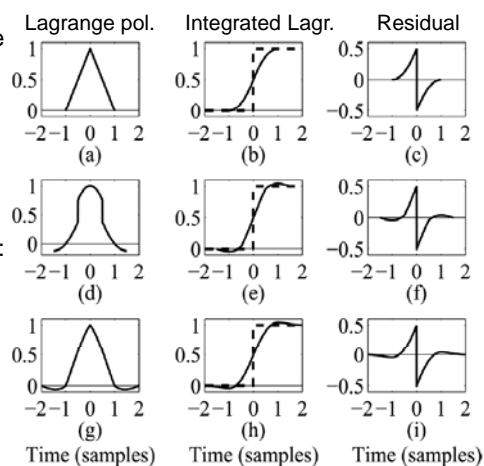
BLEP Method Example

- A shifted and sampled BLEP residual is added onto each discontinuity
- The shift is the same as the fractional delay of the step
- The BLEP residual is inverted for downward steps
(Välimäki *et al.*, 2012)



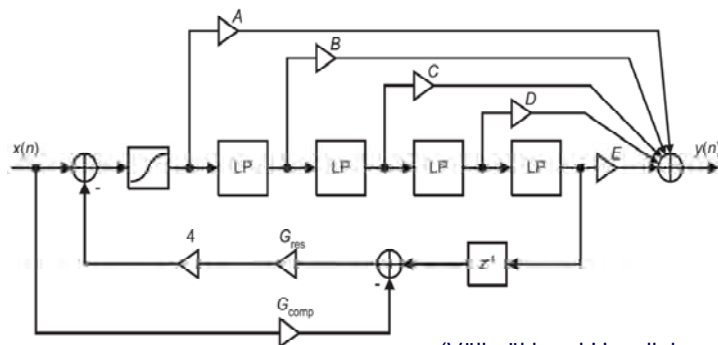
Polynomial BLEP Method

- The BLEP residual table can be replaced with a polynomial approximation (Välimäki *et al.*, 2012)
- Lagrange polynomials can be integrated and used for approximating the sinc function
- Low-order cases are of interest:
 - $N = 1$ (Välimäki and Huovilainen, 2007)
 - $N = 2$ (Välimäki *et al.*, 2012)
 - $N = 3$ (Välimäki *et al.*, 2012)

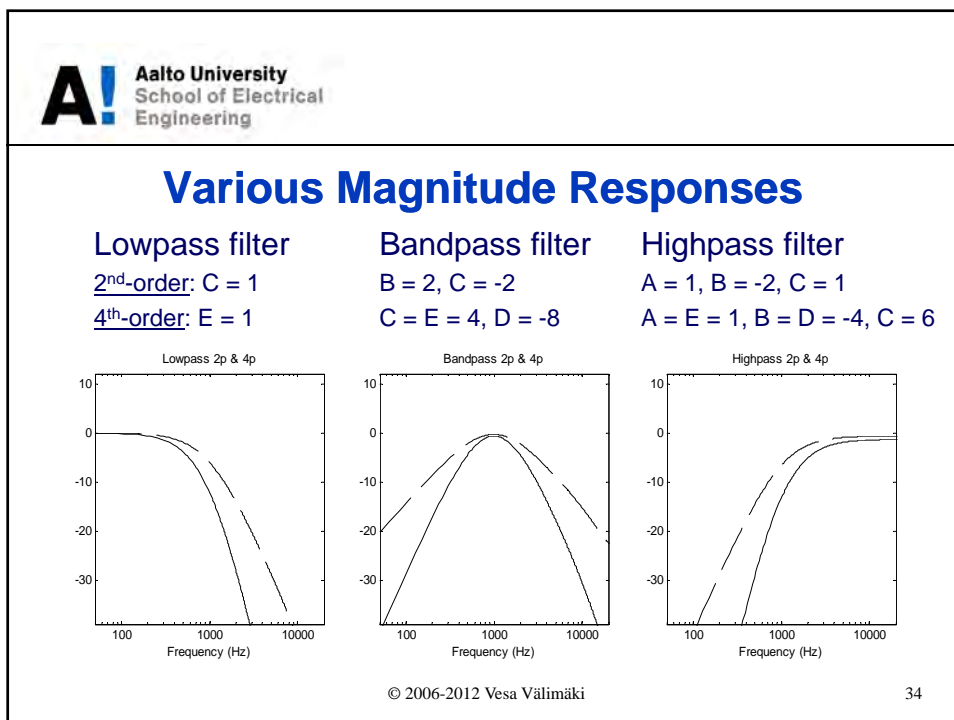
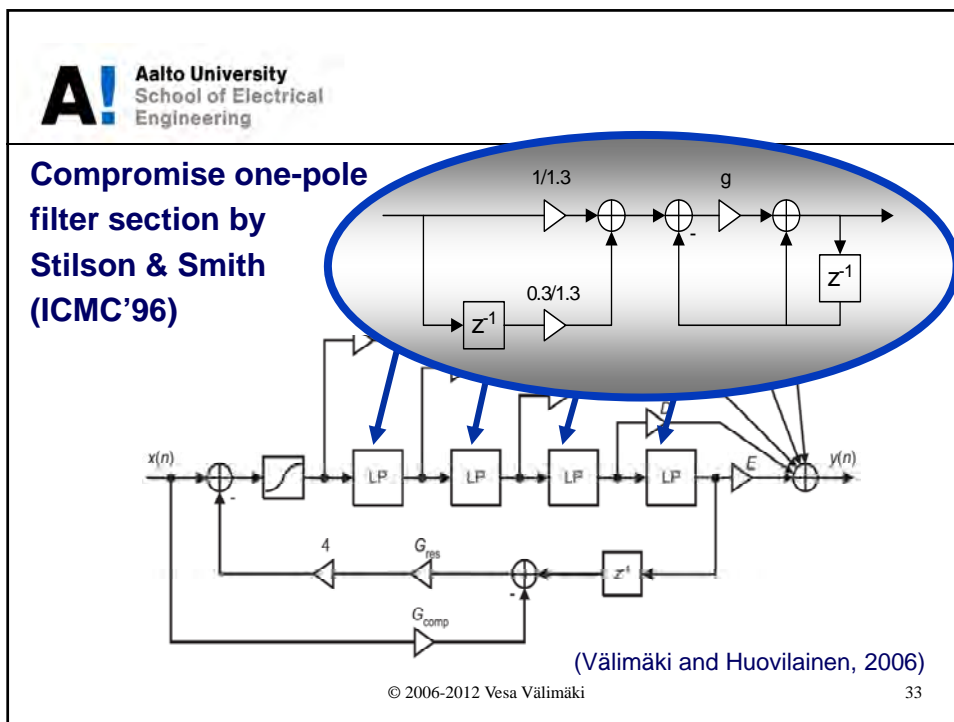


A Digital Resonant Filter

- Simplified version of the digital 4th-order Moog ladder filter (Huovilainen, DAFx 2004)

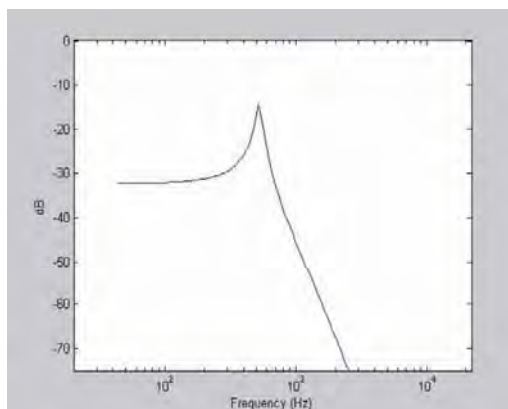


(Välimäki and Huovilainen, 2006)



Sweeping the Resonance Frequency

- Changing the resonance frequency does not affect the Q value (much)



Video by Oskari Porkka & Jaakko Kestilä, 2007

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Sweeping the Resonance Frequency

- Changing the resonance frequency does not affect the Q value (much)

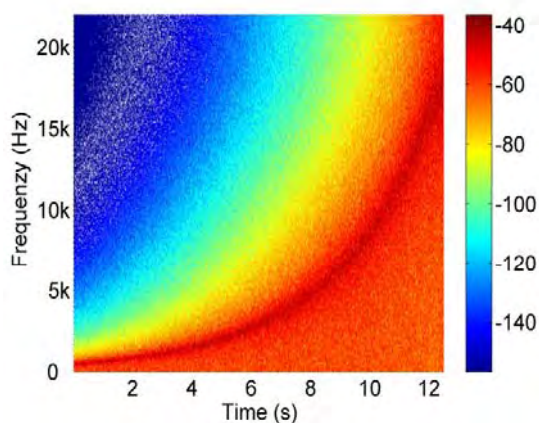


Image by Oskari Porkka & Jaakko Kestilä, 2007

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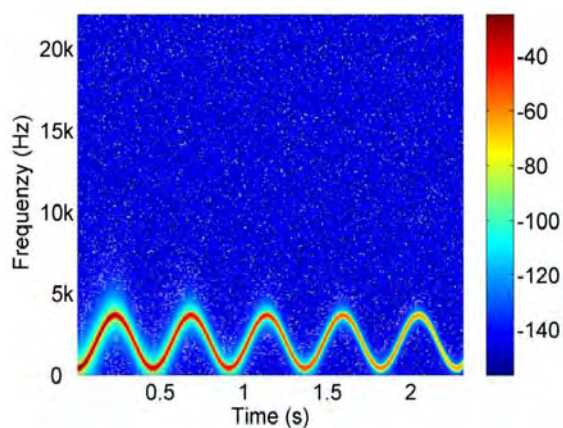
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Self-Oscillation

- When $C_{res} = 1$, the digital Moog filter oscillates for some time



- Note that C_{res} can be made larger than 1, because the TANH nonlinearity limits the amplitude and guarantees stability



Sound & image by Oskari Porkka & Jaakko Kestilä, 2007

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Moog Filter Sound Examples

Original



Lowpass with LFO,
resonance = 0.99



Original



Lowpass with sweep,
resonance = 0.8



Sounds by Oskari Porkka & Jaakko Kestilä, 2007

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Mobile Audio Processor

- Collaboration of TKK Acoustics Lab and VLSI Solution Oy (Tampere, Finland): SP-Mini project
- For mobile audio applications: phones, games, toys
- Uses the Scalable Polyphony MIDI (SP-MIDI) specification
 - A version of MIDI for mobile applications: reduced sound set, drop voices when necessary
- Main synthesis principle: digital subtractive synthesis
 - DPW oscillator algorithm used for most sounds
 - Virtual analog resonant filter (by Antti Huovilainen)
- Music examples from software simulation of the chip:
 - SP-MIDI files taken from the Web



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