Embedded Virtual Analog Modelling with Recurrent Neural Networks

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Virtual Analog Modelling

Creating a digital emulation of a classic analog audio effects.

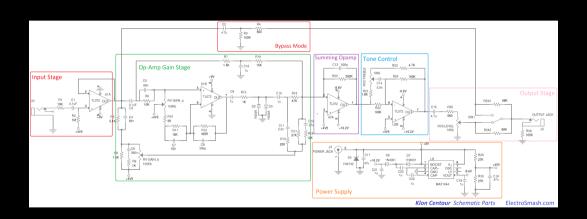
- Provide access to effects that are old or rare.
- Lower cost.
- Convenience.
- Improved understanding.

Klon Centaur

Guitar pedal made by Bill Finnegan (MIT) from 1994-2000

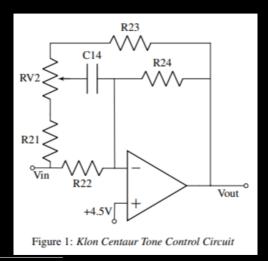


Klon Centaur Circuit Schematic



Traditional Circuit Modelling

Nodal Analysis¹



¹Smith, Physical Audio Signal Processing; Maby, Solid State Electronics.

Nodal Analysis

Laplace domain transfer function

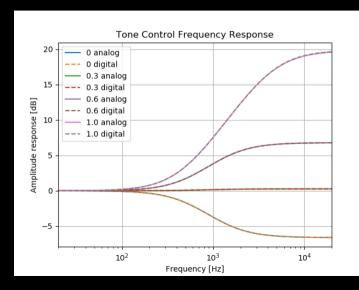
$$\frac{V_{out}(s)}{V_{in}(s)} = \frac{\frac{C_{14}\left(\frac{1}{R_{22}} + \frac{1}{R_{21} + R_{v2b}}\right)s + \frac{1}{R_{22}}\left(\frac{1}{R_{21} + R_{v2b}} + \frac{1}{R_{23} + R_{v2a}}\right)}{\frac{C_{14}\left(\frac{1}{R_{23} + R_{v2a}} + \frac{1}{R_{24}}\right)s + \frac{-1}{R_{24}}\left(\frac{1}{R_{21} + R_{v2b}} + \frac{1}{R_{23} + R_{v2a}}\right)}}$$
(1)

Bilinear transform (or other conformal map)

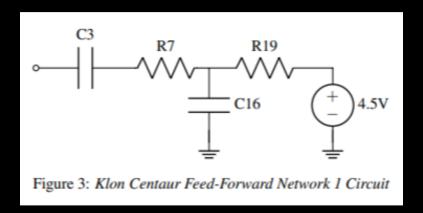
$$s \leftarrow \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}} \tag{2}$$

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Nodal Analysis



Wave Digital Filters (WDFs)2

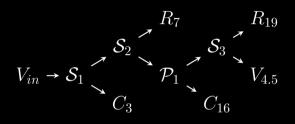


²Fettweis, "Wave digital filters: Theory and practice"; Werner, "Virtual Analog Modeling of Audio Circuitry Using Wave Digital Filters".

Wave Digital Filters (WDFs)

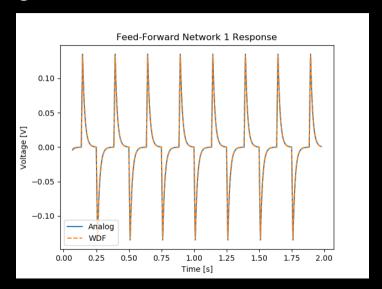
- Use wave variables instead of voltage/current.
- Port resistance free parameter.

• Discretize each circuit element separately.



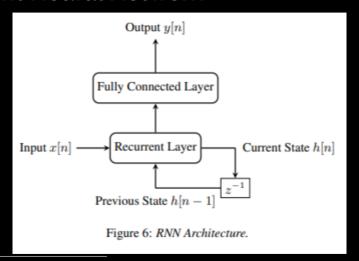
WDF tree for the Klon Centaur Feed-Forward Network 1 Circuit. S and P nodes refer to series and parallel adaptors respectively.

Wave Digital Filters (WDFs)



Neural Network Circuit Modelling

Recurrent Neural Network³



³Wright, Damskagg, and Valimaki, "Real-Time Black-Box Modelling with Recurrent Neural Networks".

Recurrent layer: Gated Recurrent Unit

$$z[n] = \sigma(W_z x[n] + U_z h[n-1] + b_z)$$
 (3)

$$r[n] = \sigma(W_r x[n] + U_r h[n-1] + b_r)$$
 (4)

$$c[n] = \tanh(W_c x[n] + r[n] \circ U_c h[n-1] + b_c)$$
 (5)

$$h[n] = z[n] \circ h[n-1] + (1-z[n]) \circ c[n]$$
 (6)

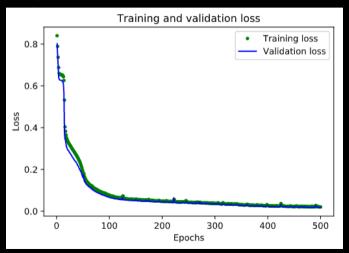
Training Data:

- $\bullet \sim 4$ minutes of guitar recordings (direct)
- Split into 0.5 second segments
- 400 training samples, 25 validation samples

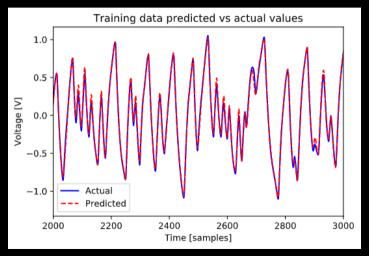
Loss Function: Error-to-Signal Ratio

$$\mathcal{E}_{ESR} = \frac{\sum_{n=0}^{N-1} |y[n] - \hat{y}[n]|^2}{\sum_{n=0}^{N-1} |y[n]|^2}$$
(7)

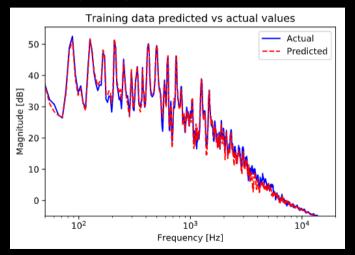
Training: 500 epochs, ~ 8 hours



Training results (time domain)



Training results (frequency domain)



Real-Time Implementation

Non-ML Implementation

- Use a combination nodal analysis, WDFs
- Control parameters for Treble, Gain, Level

ML Implementation

 RNN model for Gain Stage, nodal analysis elsewhere

 Fade between models for variable Gain control

 Custom GRU and Dense layer implementations in C++

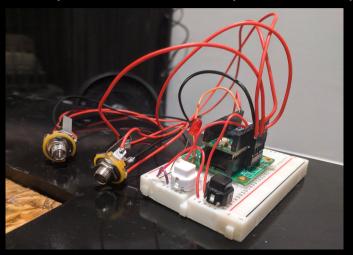
Inference Engine

- Why use a custom implentation?
 - GRU support in TFLite is still experimental
 - Real-time audio thread: no locking, no memory allocation
- Currently using STL functions (std::inner_product)
- Eventually upgrade to Eigen or TFLite

Desktop Audio Plugin (JUCE/C++)



Teensy 4.0, Teensy Audio Shield, Teensy Audio Library



Results: Performance

Compute time per second of audio.

Block Size	NonML Speed	ML Speed
8	0.0723437	0.0528792
16	0.0703079	0.0510437
32	0.0652856	0.0511147
64	0.0662835	0.0502434
128	0.0666593	0.0495194
256	0.0696844	0.0480298
512	0.0669037	0.0477946
1024	0.060816	0.0488841
2048	0.0695175	0.0488309
4096	0.0623839	0.0472191

Results: Summary

- Subjectively, non-ML and ML models sound very similar.
- ML model has slightly damped high frequency response, (not a big deal on guitar input; more noticeable on other audio).
- ML model is more efficient!

Live Performance!