



Professur
Allgemeine Nachrichtentechnik

Prof. Dr.-Ing. Udo Zölzer



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UNIVERSITÄT

Parameters Automation of Dynamic Range Compressor

Literature Review - Project Work

Aritra Mazumdar

20 December 2018

- Introduction
- Theory
- Automation
- Approach 1
- Approach 2
- Discussions
- Conclusion

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Introduction

Dynamic Range Compressor

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Motivation:

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- Project work involved building DRC application
- Application accepts static set of parameters
- Adapting parameters to signal dynamics gives better results

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Dynamic Range of a signal can be controlled using a process named as **Dynamic Range Compressor**.

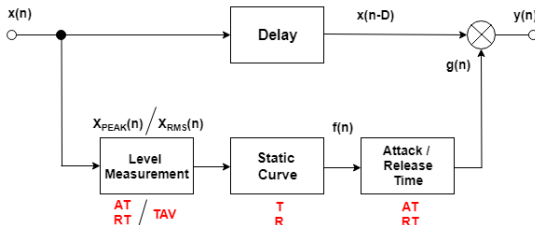
Motivation:

- Project work involved building DRC application
- Application accepts static set of parameters
- Adapting parameters to signal dynamics gives better results
- Compressor with automated parameters is better approximation

- Introduction
- **Theory**
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Theory

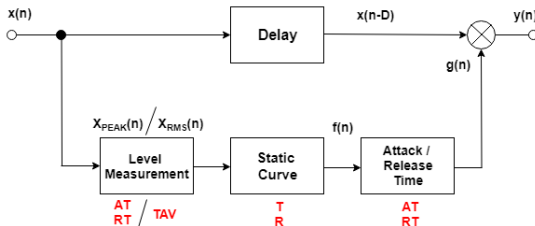
Block Diagram



The figure above represents a basic Dynamic Range Compressor system.

Theory

Block Diagram



The figure above represents a basic Dynamic Range Compressor system.

$$y(n) = x(n - D) \cdot g(n) \quad (1)$$

- Threshold (T)

- Threshold (T)
- Ratio (R)

- Threshold (T)
- Ratio (R)
- Knee Width (W)

- Threshold (T)
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- Attack Time (τ_A)

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- Release Time (τ_R)
- Make-up Gain (g)

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Automation

Goals and Approaches

- Based on signal

Automation

Goals and Approaches

- Based on signal
 - Calculate signal metrics

Automation

Goals and Approaches

- Based on signal
 - Calculate signal metrics
 - Find a relation between calculated metrics and Compressor parameters

Automation

Goals and Approaches

- Based on signal
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- Based on Supervised Learning

Automation

Goals and Approaches

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 - Music similarity measure between reference and processed audio

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Approach 1

Attack and Release Times

- **Crest Factor Method:**

Approach 1

Attack and Release Times

- **Crest Factor Method:**
 - Ratio of Peak to RMS signal level

Approach 1

Attack and Release Times

- **Crest Factor Method:**
 - Ratio of Peak to RMS signal level
 - Low for steady state signal

Approach 1

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 - **Calculation:**

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$$y_C[n] = y_{PEAK}[n]/y_{RMS}[n] \quad (2)$$

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- **Calculation:**

$$y_C[n] = y_{PEAK}[n] / y_{RMS}[n] \quad (2)$$

$$\tau_A[n] = 2 \cdot \tau_{Amax}[n] / y_C^2[n] \quad (3)$$

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$$\tau_R[n] = 2 \cdot \tau_{Rmax}[n] / y_C^2[n] - \tau_A[n] \quad (4)$$

Approach 1

Attack and Release Times (contd..)

- **Spectral Flux Method:**

Approach 1

Attack and Release Times (contd..)

- **Spectral Flux Method:**
 - Rapidity in change of power spectrum of signal

Approach 1

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- More sensitive than Crest Factor

Approach 1

Attack and Release Times (contd..)

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Approach 1

Attack and Release Times (contd..)

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- Rapidity in change of power spectrum of signal
- Low for steady state signal
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- **Calculation:**

$$SF(n) = \frac{\sum_{k=-N/2}^{N/2-1} H(|X(n, k)| - |X(n-1, k)|)}{\sum_{k=-N/2}^{N/2-1} |X(n, k)|} \quad (5)$$

Approach 1

Attack and Release Times (contd..)

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$$SF_{smooth}[n] = \max(x[n], \alpha \cdot SF_{smooth}[n-1] + (1 - \alpha) \cdot SF[n]) \quad (6)$$

Approach 1

Attack and Release Times (contd..)

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Approach 1

Attack and Release Times (contd..)

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$$\tau_R[n] = 2 \cdot \tau_{Rmax}[n] / SF_{smooth}^\gamma[n] - \tau_A[n] \quad (8)$$

Approach 1

Attack and Release Times - Evaluation

- Subjective evaluation based on human preference of parameters

Approach 1

Attack and Release Times - Evaluation



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- Subjective evaluation based on human preference of parameters
- Evaluation results compared with automated parameters

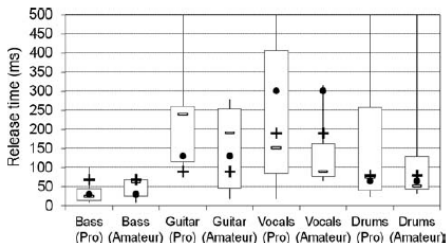
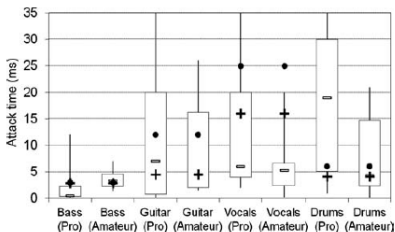
Approach 1

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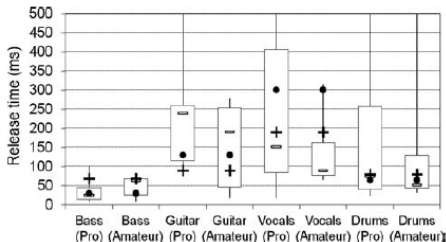
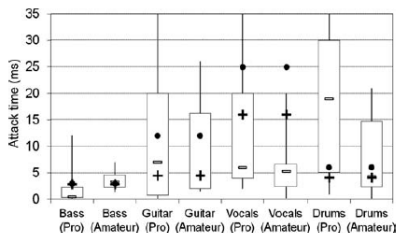
Approach 1

Attack and Release Times - Evaluation



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- (-) Median value of response
- (●) Crest Factor Automation
- (+) Spectral Flux Automation

Approach 1

Threshold and Ratio

- **Threshold:**

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 - Defines desired compression

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Threshold and Ratio

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- **Ratio:**
 - To be set to infinity

Approach 1

Threshold and Ratio

- **Threshold:**

- Defines desired compression
- To be controlled by user

- **Ratio:**

- To be set to infinity
- Soft Knee with variable width equivalent to automatic ratio

Approach 1

Knee Width

- **Basic Method:** Bases on gain reduction of input signal

Approach 1

Knee Width

- **Basic Method:** Bases on gain reduction of input signal

$$g_{Dev}[n] = \alpha \cdot g_{Dev}[n - 1] + (1 - \alpha) \cdot (g[n] - g_{Est}) \quad (9)$$

Approach 1

Knee Width

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$$k = \begin{cases} 0.6, & SF_{min,avg} > 0.1 \\ 0.05, & SF_{min,avg} \leq 0.1 \end{cases} \quad (14)$$

Approach 1

Knee Width - Evaluation

- Subjective evaluation based on human preference of parameters

Approach 1

Knee Width - Evaluation

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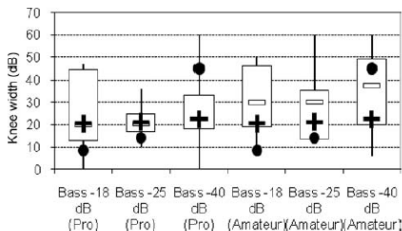
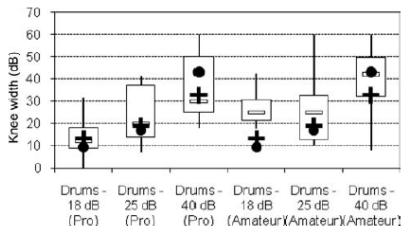
Approach 1

Knee Width - Evaluation



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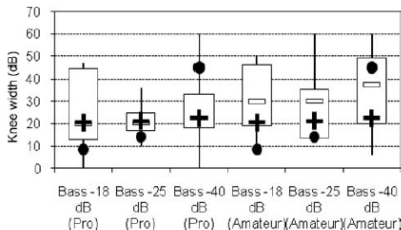
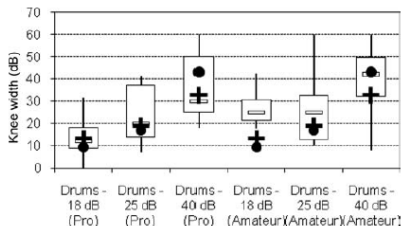
Approach 1

Knee Width - Evaluation



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- Subjective evaluation based on human preference of parameters
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- (-) Median value of response
- (●) Gain Reduction Dependent Automation
- (+) Signal Information Dependent Automation

Approach 1

Make-up Gain

- **Compression based Method:**

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- **Loudness based Method:**

- Compares perceived loudness before and after compression

■ Compression based Method:

$$g_{make-up}[n] = -(g_{Dev}[n] + g_{Est}) \quad (15)$$

■ Loudness based Method:

- Compares perceived loudness before and after compression
- Calculates make-up gain from loudness difference

Approach 1

Make-up Gain - Evaluation

- Subjective evaluation based on human preference of parameters

Approach 1

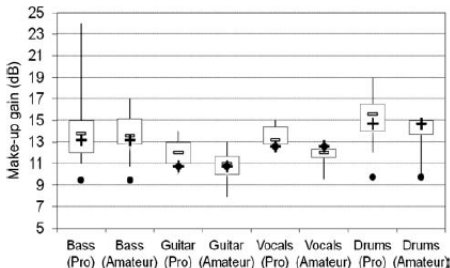
Make-up Gain - Evaluation

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Approach 1

Make-up Gain - Evaluation

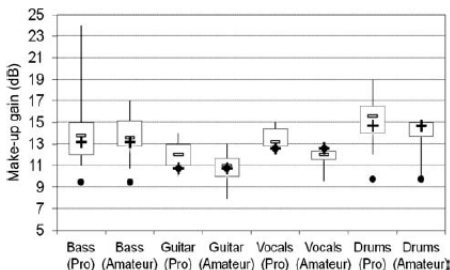
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Approach 1

Make-up Gain - Evaluation

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- (-) Median value of response
- (●) Compression based Automation
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Approach 2

Audio Features Extraction



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$$SC_{mean} = E\left[\frac{\sum_{k=0}^{K-1} k \cdot Y(n, k)}{\sum_{k=0}^{K-1} Y(n, k)}\right], Y(n, k) = |X(n, k)| \quad (16)$$

Approach 2

Audio Features Extraction



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$$SC_{mean} = E\left[\frac{\sum_{k=0}^{K-1} k \cdot Y(n, k)}{\sum_{k=0}^{K-1} Y(n, k)}\right], Y(n, k) = |X(n, k)| \quad (16)$$

$$SC_{var} = Var\left[\frac{\sum_{k=0}^{K-1} k \cdot Y(n, k)}{\sum_{k=0}^{K-1} Y(n, k)}\right] \quad (17)$$

Approach 2

Audio Features Extraction



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$$SV_{mean} = E[\sqrt{E[Y(n, k)^2] - (E[Y(n, k)])^2}] \quad (18)$$

Approach 2

Audio Features Extraction



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Approach 2

Audio Features Extraction



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$$RMS_{mean} = E\left[\sqrt{\frac{1}{M} \cdot \sum_{m=0}^{M-1} x(m)^2}\right] \quad (20)$$

Approach 2

Audio Features Extraction



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$$RMS_{var} = Var\left[\sqrt{\frac{1}{M} \cdot \sum_{m=0}^{M-1} x(m)^2}\right] \quad (21)$$

Approach 2

Regression Model - Training

- **Training Data:** Violin Samples (RWC isolated note database)

Approach 2

Regression Model - Training



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- **Training Data:** Violin Samples (RWC isolated note database)
- **Training Set:**

Training sets (size)	Conditions			
	Thr(dB)	Ratio	Att(ms)	Rel(ms)
A (60*50)	0:1:49	2	5	200
B (60*50)	37.5	1:0.4:20	5	200
C (60*100)	37.5	2	1:1:100	200
D (60*100)	37.5	2	5	50:10:1000

Approach 2

Regression Model - Training

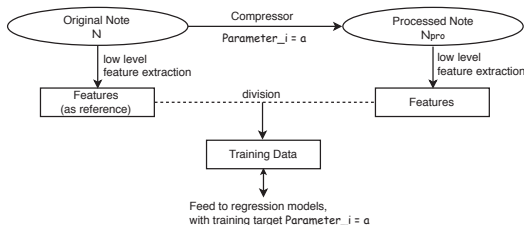


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- **Training Data Generation:**



Approach 2

Regression Model - Training

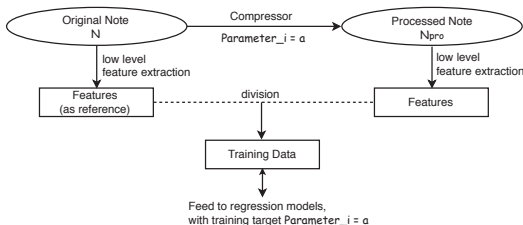


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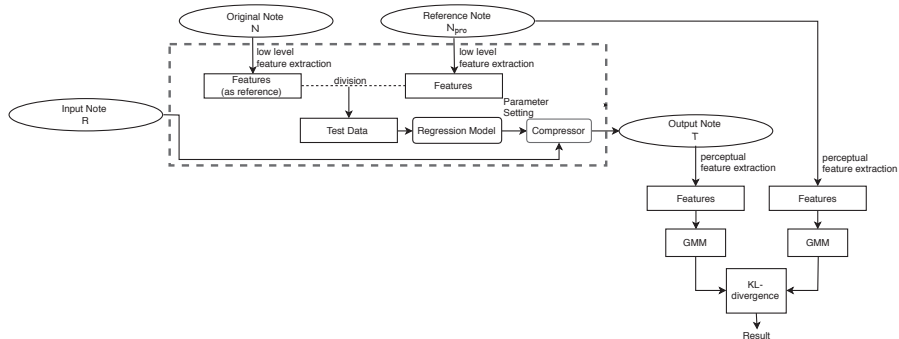
- **Regression Models:** Linear and Random Forest Regression

Approach 2

Automation Process



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Approach 2

Similarity Measure

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Approach 2

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Approach 2

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Approach 2

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Violin	Threshold	Ratio	Attack	Release
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$D(N_{pro}, T)_{LR}$	19.799	20.911	22.852	20.994
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- Theory
- Automation
- Approach 1
- Approach 2
- **Discussions**
- Conclusion

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- Better results with approaches using Spectral Flux, SF based Optimization and Loudness

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- Variation of preference of ballistics across styles

- Include additional audio features mentioned in Approach 1 to train Regression Model in Approach 2 for better results

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- Evaluation on more complex tracks

1. Dimitrios Giannoulis, Michael Massberg, Joshua D. Reiss, "Parameter automation in a dynamic range compressor", J. Audio Eng. Soc., Vol. 61, No. 10, October 2013
2. Di Sheng, György Fazekas , "Automatic control of the dynamic range compressor using a regression model and a reference sound", Proceedings of the 20th International Conference on Digital Audio Effects (DAFx-17), Edinburgh, UK, September 5-9, 2017
3. Gary Bromham, David Moffat, Mathieu Barthet, György Fazekas , "The impact of compressor ballistics on the perceived style of music", Audio Engineering Society, Convention Paper 10080, presented at the 145th Convention New York, NY, USA, October 17-20, 2018

Conclusion

Thank You

Thanks for your time

Any Question or Suggestion?