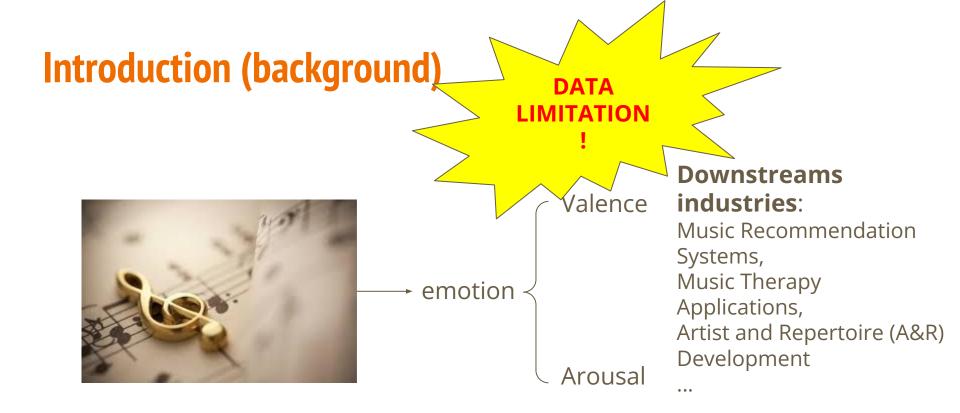
Filters Implementation in Data Augmentation on Music Emotion Prediction

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Augmentation

- Pitch shift
- Add background noise
- Reverb
- <u>Filters</u>

Lowpass

Highpass

Bandpass

Peaking

Low shelf

High shelf

Butterworth
Chebyshev (type I)
Chebyshev (type II)
Elliptic

Dataset: PMEmo

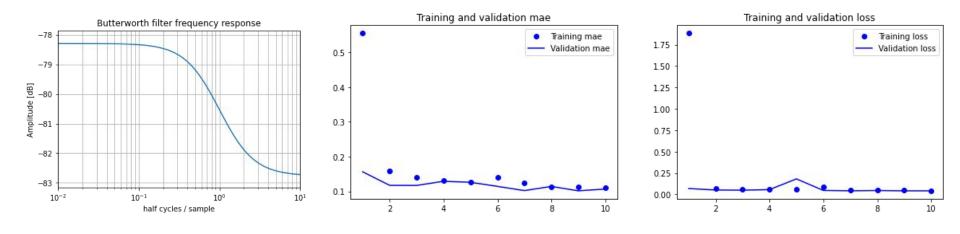
includes 794 songs with both static and dynamic emotion labels (valence & arousal), chorus clips in MP3 format

Baseline results

	Test Loss (Mean squared error)	Test MAE (Mean absolute error)
Original	0.036	0.160
Pitch shift	0.032	0.146
Background noise	0.034	0.157
Reverb	0.031	0.145

Filters (lowpass)

Butterworth filter: (order = 4, frequency gain drop = 1000Hz)

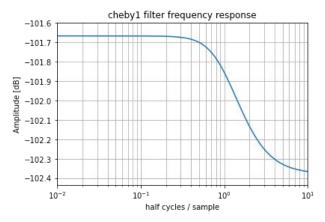


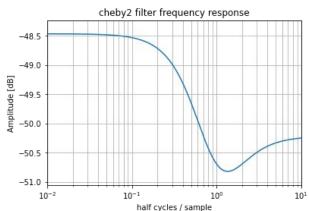
Test Loss: 0.04165396839380264, Test MAE: 0.10684800893068314

Filters (lowpass)

Chebyshev (type I) filter: (order = 4, maximum ripple allowed below unity gain in the passband = 5dB)

Chebyshev (type II) filter: (order = 4, minimum attenuation required in the stop band = 50dB)





Chebyshev I:

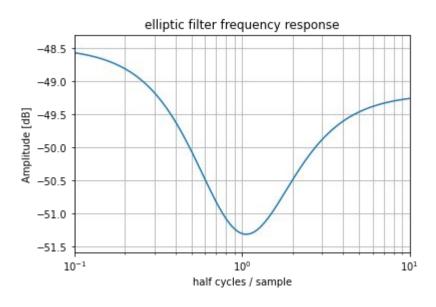
Test Loss: 0.05766607075929642, Test MAF: 0.12264931201934814

Chebyshev II:

Test Loss: 0.06217707321047783, Test MAF: 0.12897001206874847

Filters (lowpass)

Elliptic filter: (order = 4, maximum ripple allowed below unity gain in the passband = 5dB, minimum attenuation required in the stop band = 50dB)

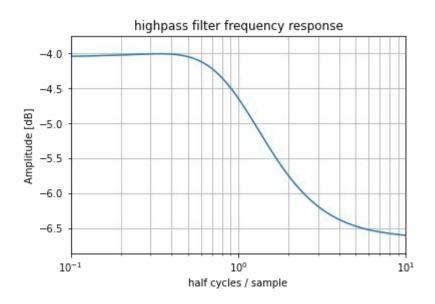


Test Loss: 0.0508585125207901,

Test MAE: 0.12048794329166412

Filters (highpass)

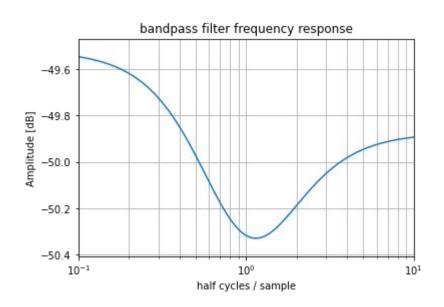
Elliptic filter (same parameters)



Test Loss: 0.05013992264866829, Test MAE: 0.1088569238781929

Filters (bandpass)

Elliptic filter



Test Loss: 0.05656576529145241, Test MAE: 0.11583404242992401

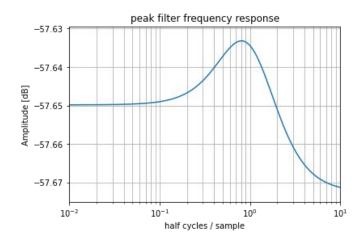
Filters (peaking)

(Frequency to be retained in a signal = 200Hz,

Quality factor -

characterizes peak filter -3 dB bandwidth bw relative to its center frequency, Q = w0/bw

=30)



Test Loss:

0.050447650253772736,

Test MAE:

0.11729744076728821

Filter (high shelf & low shelf)

min_center_freq=100, max_center_freq=1000

High shelf:

$$H(s) = A \frac{As^2 + \frac{\sqrt{A}}{Q}s + 1}{s^2 + \frac{\sqrt{A}}{Q}s + A}$$

Low shelf:

$$H(s) = A \frac{s^2 + \frac{\sqrt{A}}{Q}s + A}{As^2 + \frac{\sqrt{A}}{Q}s + 1}$$

Test Loss:

0.08168621361255646,

Test MAE:

0.1257268637418747

Test Loss:

0.05099983885884285,

Test MAE:

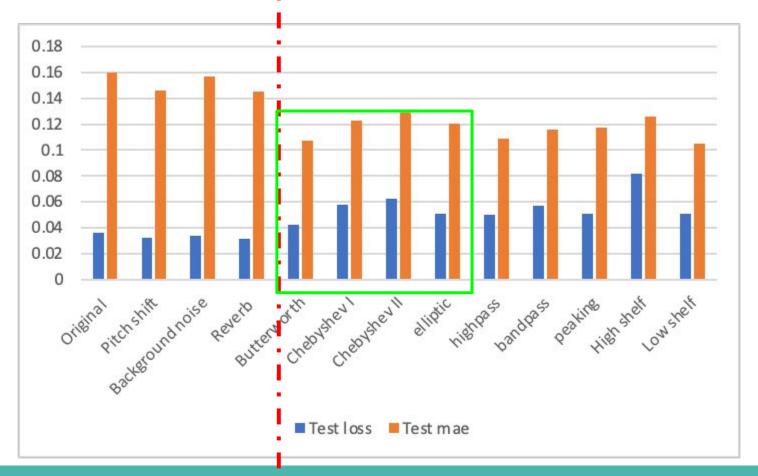
0.10496504604816437

results

lowpass

		Test loss	Test mae
	butterworth	0.04165396839380264	0.10684800893068314
	Chebyshev I	0.05766607075929642	0.12264931201934814
	Chebyshev II	0.06217707321047783	0.12897001206874847
	elliptic	0.0508585125207901	0.12048794329166412
	highpass	0.05013992264866829	0.1088569238781929
	bandpass	0.05656576529145241	0.11583404242992401
	peaking	0.05044765025377273	0.11729744076728821
	High shelf	0.08168621361255646	0.1257268637418747
	Low shelf	0.05099983885884285	0.10496504604816437

results



conclusion

$$ext{MSE} = rac{1}{n} \sum_{i=1}^n (y_i - \hat{y}_i)^2.$$

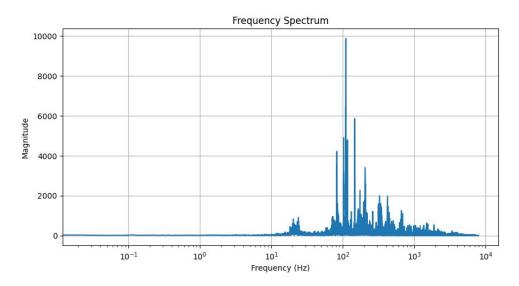
$$ext{MAE} = rac{1}{n} \sum_{i=1}^n |y_i - \hat{y}_i|.$$

MSE is more sensitive to outliers. MSE might be more suitable for applications where it is critical to penalize large errors more heavily and when the data is relatively free of outliers.

- 1. Using filters to do the augmentation leads to some notable outliers, but the overall performances are better.
- 2. All of filters contributes to the improvement. Butterworth lowpass filter performs the best.
- 3. The results of applying low-pass filters have no commonality. The results depend on the specific situations.

Future Thoughts

The filters can be designed based on audio's characteristics dynamically.



Combination of different filters.

Thank you!