



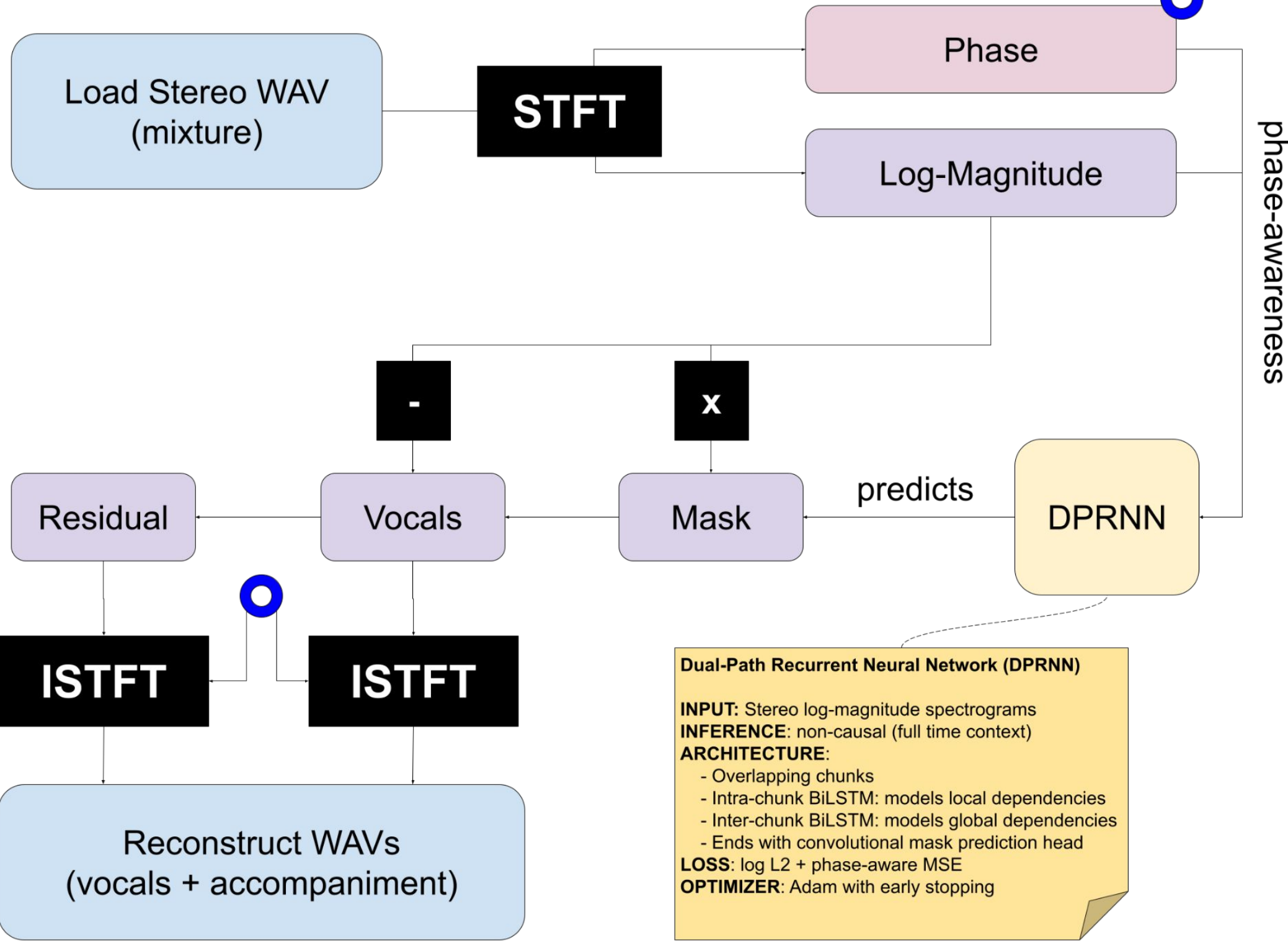
Objectives

- **Build and evaluate** a **deep learning system** to **separate components of an audio track**: vocals and accompaniment.
- **Compare** our developed model with a popular **pre-trained model (Open-Unmix UMXHQ)** through the use of standard **separation metrics** that evaluate the quality of the separation as well as the lack of unwanted sources and distortions.

Problem

- **Audio source separation** is the process of isolating specific sounds from an original audio source containing a mixture of multiple elements, namely instruments and vocals.
- For example, we might want to separate the instrumental of a song from its vocals to make a karaoke track or isolate a specific instrument from a song so a musician can learn that part.
- In this type of task, we usually have 2 main models:
 1. **Time-Frequency Domain**, using *STFT* to separate sources based on magnitude/phase or by predicting a mask to isolate target sources from the mixture.
Drawbacks: Phase reconstruction can limit quality; resolution trade-offs (time vs. frequency).
 2. **Time-Domain**, by learning to map raw waveforms directly to separated sources (*end-to-end*).
Drawbacks: Often more complex, require more training data and compute resources.

Methodology



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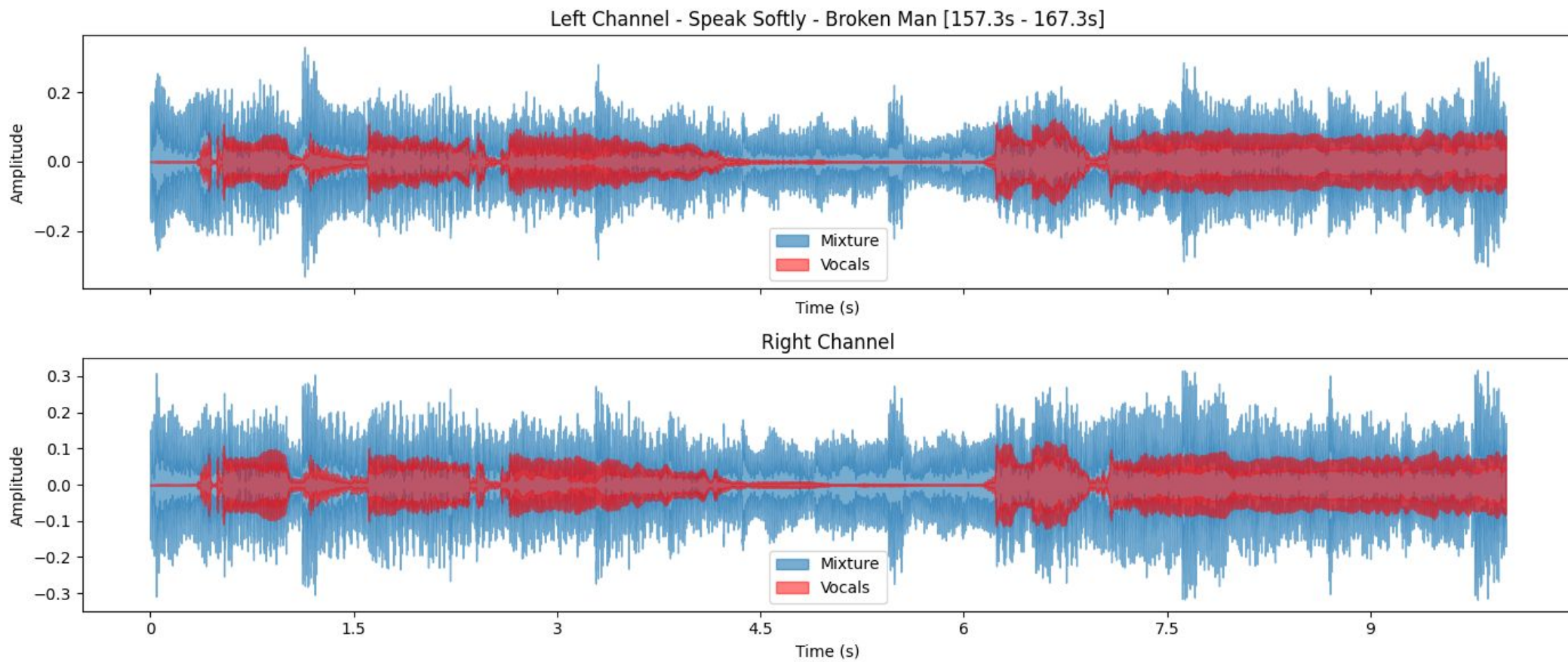
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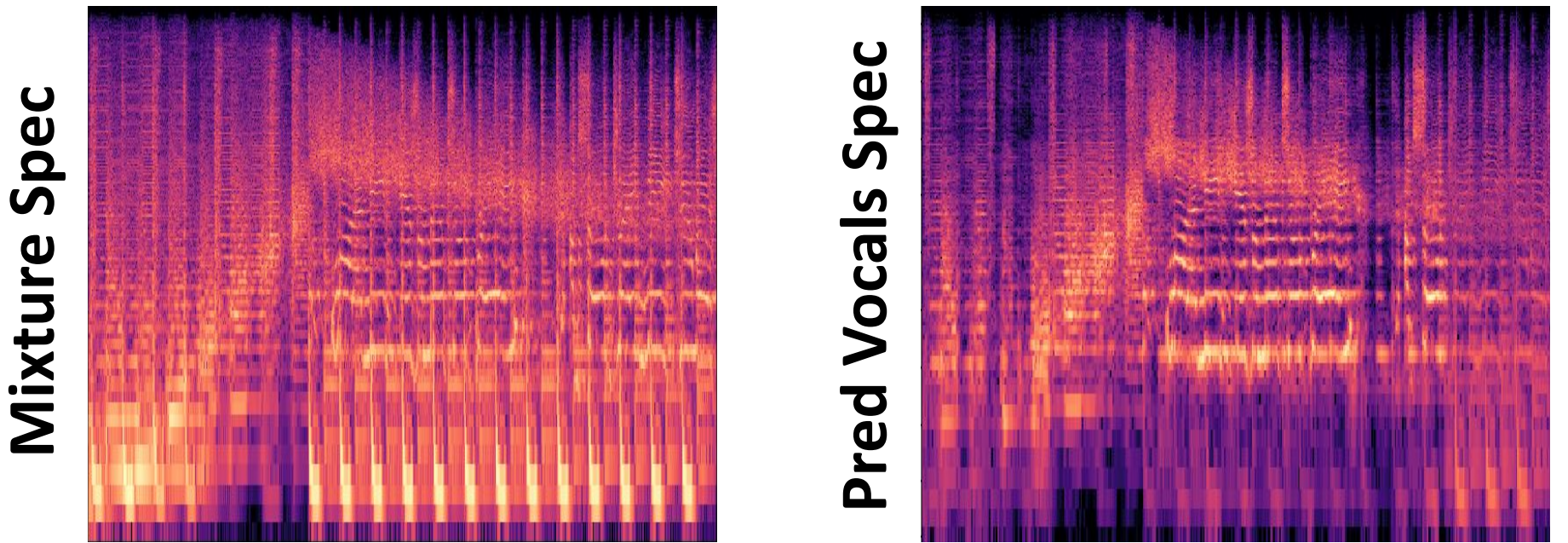
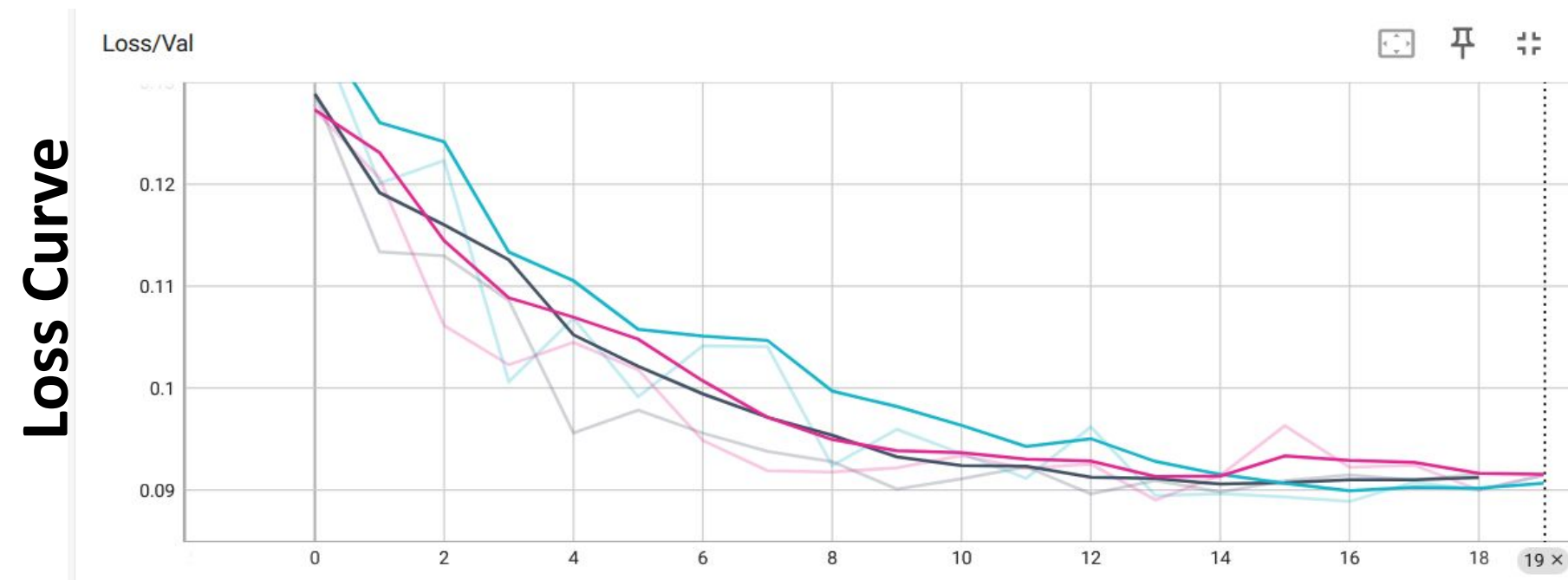
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Datasets

- **MUSDB18-HQ**, an industry standard dataset, was used.
 - 150 full-length **stereo** tracks (~10 hours) of different genres split into 86 train tracks, 50 test tracks and 14 validation tracks.
 - Each track comes pre-split into **mixture**, **bass**, **drums**, **vocal** and **other** .wav files, of which we used the **vocal** and **mixture** files.
- Example **waveform plot** of one of the tracks (10 seconds):



Results



- Comparison of industry standard metrics between the **2 models**, tested on **30 second** chunks of **25 random** tracks:

Target	Model	SDR	SIR	ISR	SAR
Vocals	<i>Open-Unmix</i>	5.259	12.508	14.385	6.607
Accompaniment	<i>Open-Unmix</i>	12.951	19.901	21.255	14.464
Vocals	<i>DPRNN-ass</i>	2.516	5.445	9.482	6.011
Accompaniment	<i>DPRNN-ass</i>	6.094	22.865	6.941	8.449

Conclusions

- Our project demonstrates that **it is suitable to use a DPRNN-based** deep learning approach in the time-frequency domain for the audio source separation task, because the **dual intra/inter chunk RNN** balances **local** and **global context**.
- Adding the **phase-aware loss** helped a lot, even if we **don't yet have results comparable to the state of the art**.
- **Future directions** could include
 - Exploring **time-domain** architectures and **multi-target sep**.
 - Exploring more advanced **data augmentation** techniques like **pitch shifting** or **time stretching** before *STFT* and **parameter tuning** with *GridSearch*