

AI Audio Enhancement Research

Project Status & Future Roadmap

A comprehensive overview of our progress in digital audio processing and AI model development

What is this project about

Can we use AI to upscale lossy audio files to near-lossless audio quality?

Benefits:

- Reduce streaming bandwidth if native to streaming devices
- Optimize disk space for downloaded audio files
- Allow lossless streaming on Bluetooth devices
- Highly distributed IoT networks for audio processing

Tools

AI

- Model - 1D U-Net architecture
 - PyTorch for ML
 - Librosa for processing
- Hosted through Kaggle

Back-End Streaming

- API:
 - FastAPI framework
 - Librosa for processing
 - Pydantic for DTO/validation
 - Uvicorn for hosting
- DB
 - MongoDB for file metadata
 - File Store for audio Files

FrontEnd

- Flutter app
 - Network packaging for API
 - Web audio playback (Audio Element)

What Went Well

AI Model Development Progress

- Advanced the AI output that has a level of **interpolation**

Streaming Platform

- Created simple streaming platform that sends File meta data and streams audio files to a client

Challenges Encountered

Dataset Limitations

- Lack of diverse datasets for comprehensive training
- Affecting interpolation due to lack of diverse files

Streaming

- Easily storing and streaming files was a challenge

Key Features:

- Mock Streaming service
- Audio playback from streaming service
- AI upscaling output with small level of interpolation

Future Plans & Roadmap

AI Model Enhancement

- Create/find data sets
- Optimize for native implementation on less powerful devices

Streaming Simulation

- Integration with AI that sends lossy files to "enhance"
- Integrate evaluation metrics from previous iterations

Key Learnings & Technical Insights

Digital Audio Processing Mastery

- Comprehensive understanding of **audio compression algorithms**
- In-depth knowledge of **perceptual audio coding**

Streaming output

- Easy streaming interface that will easily display differences between files

Sprint 1 overview

- Week 1: Defined **terminology** - began **audio evaluation**
- Week 2: **More terminology** - Continued evaluation - **audio I/O** - Began **AI model development**
- Week 3: ***Even more* terminology** - Meta-data handling. Differentiating data sets
- Week 4: **Surprise! more terminology** - Graph Construction/Modeling - **AI training**

Sprint 2 overview

- Week 5: **Data set and AI refactoring**
- Week 6: Began creating **DB** and **streaming API**.
- Week 7: Finished **DB and backend**. Began creating **frontend flutter interface**
- Week 8: Fulling integrating **Streaming Playback** to front end with **Meta-Data**

Metrics

- Total individual Lines of Code (LoC): ~4085
- Number of individual features completed: 5
 - Total Features: 8
- Number of individual requirements completed: 11
 - Total requirements 24
- Individual burndown rate (%): 45%

Next Steps & Immediate Actions

Sprint ???

1. Acquire **MORE** datasets from diverse sources
2. Move AI model hosting to **Google Colab** for **GPU** usage
3. **Integrate API to Model for output to streaming**
4. **Integrate evaluation metrics to the frontend**
5. **Create Native instances of AI Model**

Technical Implementation Overview

Backend Architecture

- Audio file processing - chunking for streaming
- Python-based processing pipeline
- Mongo-DB integration with file-store
- API for managing files

Analysis Components

- Flutter/dart front end implementation
- Audio preprocessing and normalization both on front-end and back-end
- User visualization/listening platform