

MixingBuddy: A Multimodal LLM for Mix Critique and Advice

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Advisor: Dr. Alexander Lerch

Fall 2025 Project Proposal



Georgia Tech · College of Design
Center for
Music Technology

Brief Introduction

overview

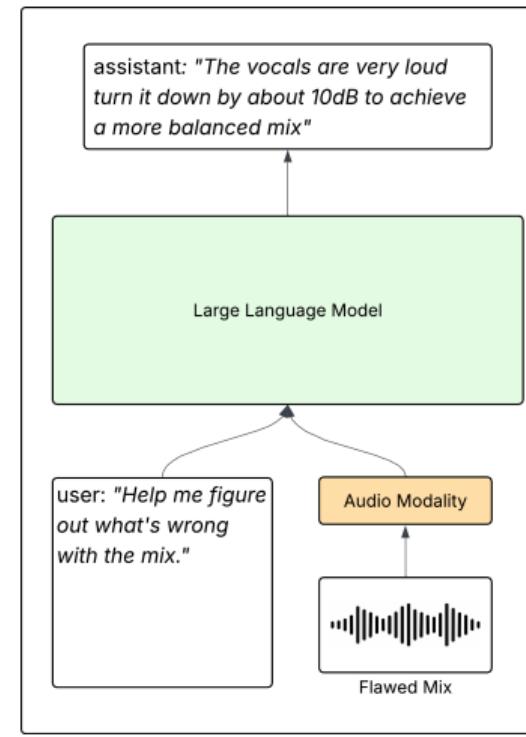
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- This research develops a **multimodal** system that leverages a **pre-trained LLM** to analyze **raw audio** and provide actionable feedback on **flawed mixes**.
- As a starting point, we focus on generating advice for **gain-balancing** only, with **audio-only** input.



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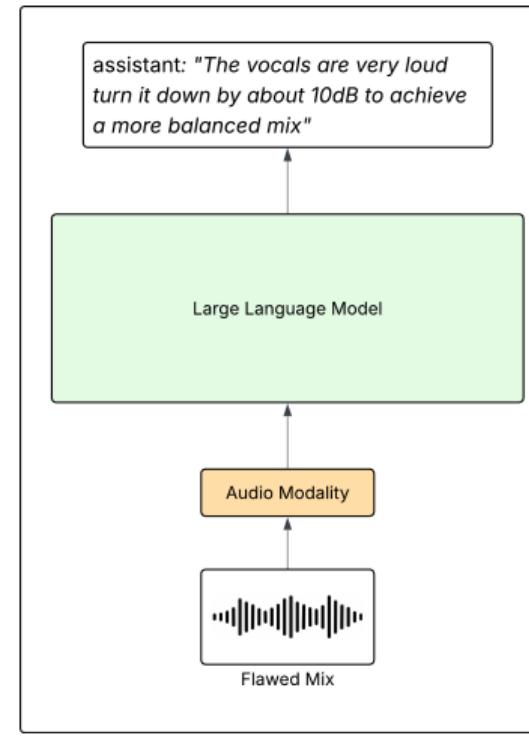
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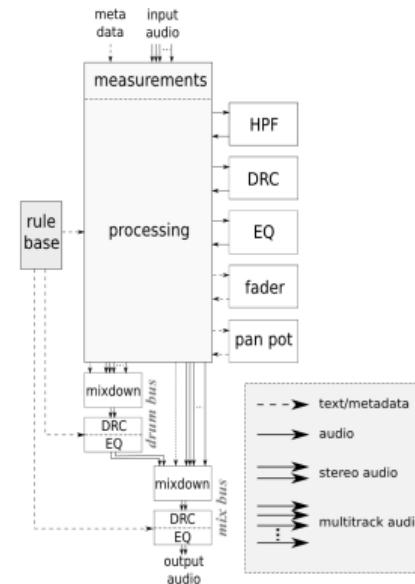
Rule-Based & Deep Learning Approaches

Rule-Based and Traditional Machine Learning Systems

- Knowledge-engineered autonomous mixing [1]
- A machine-learning approach for instrument-specific application of artificial reverberation. [2]

Deep Learning Architectures

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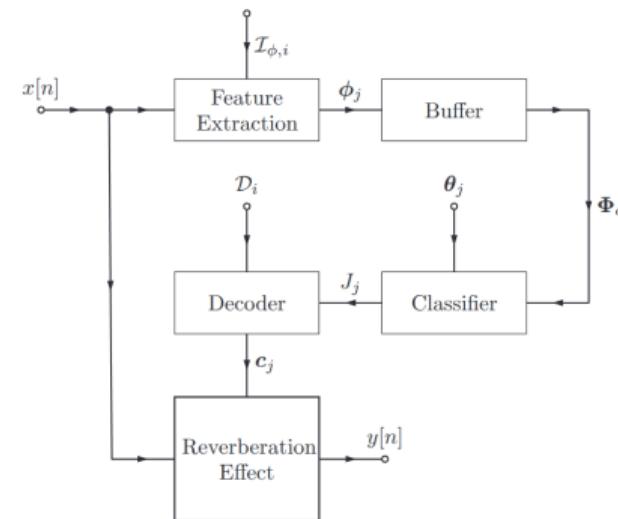


Fig. 1. Reverb application.

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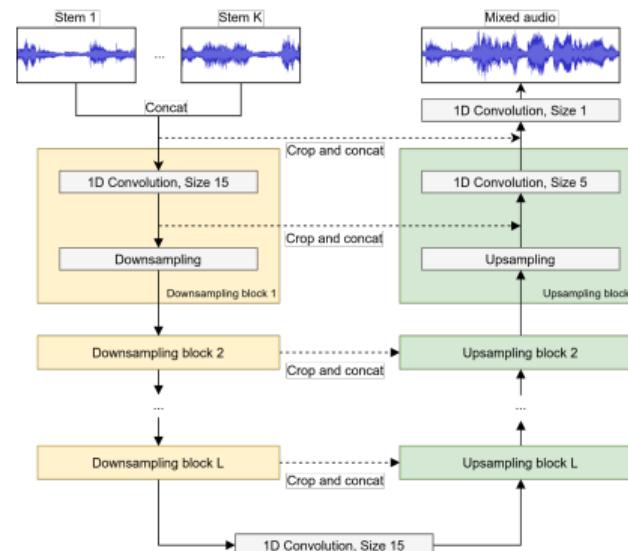
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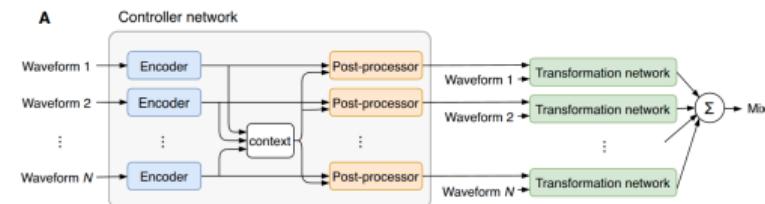
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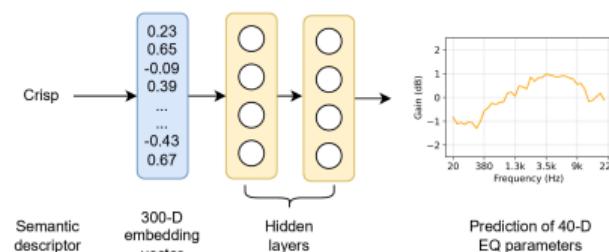


Automatic Mixing Review

Semantic Approaches

Language-Audio Integration

- Word-embedding approaches linking language and audio for audio effect parameter recommendations [5], [6], [7]
- Prompt-driven interfaces mapping natural language to mixing tasks [8], [9], [10]

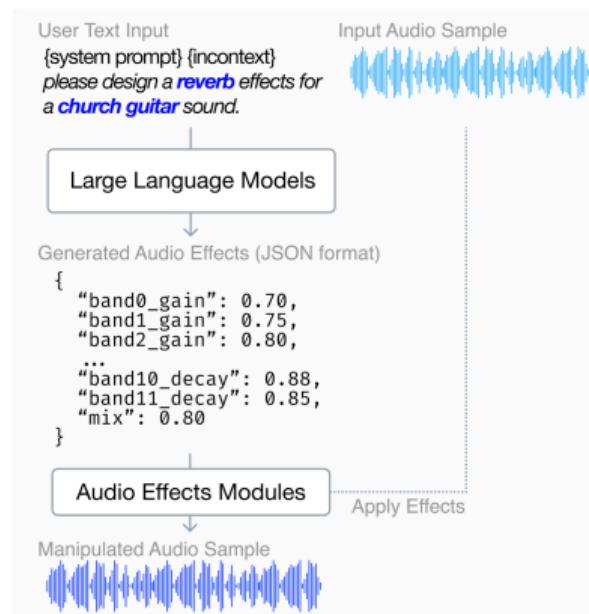


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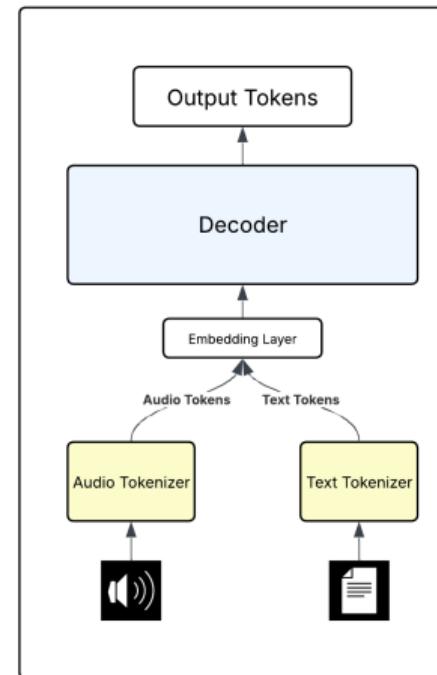
Architectural Approaches for Audio-Language Models

■ Direct Tokenization (Unified Approach):

converts raw audio into discrete tokens via audio codecs. the LLM vocabulary is extended to include audio tokens [11], [12], [13].

■ Feature Extraction (Cascade Approach):

uses audio-specific encoders/decoders with the LLM as a central backbone (e.g., LTU) [14], [15].

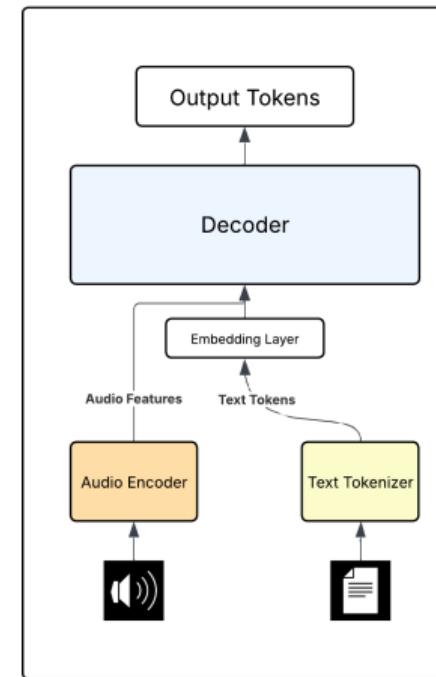


Unified Approach

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Cascade Approach

Research Questions

primary

Primary Research Question

- To what extent can an **Audio-Language Model** learn the **relative gain** relationships among multitrack stems and generate musically effective gain-balancing **advice**?

Research Questions

secondary

Secondary Questions

- **Model Understanding:** What model architecture best represents and reasons about the input mix for learning relative gain relationships?
- **Mixing Conventions:** To what extent does the model's advice reflect established mixing conventions?
- **Communication:** How effectively does the model communicate its advice in a way that is clear, actionable, and “correct”?
- **Usefulness:** How do audio engineers and producers evaluate the effectiveness, musicality, and real-world usefulness of the advice?

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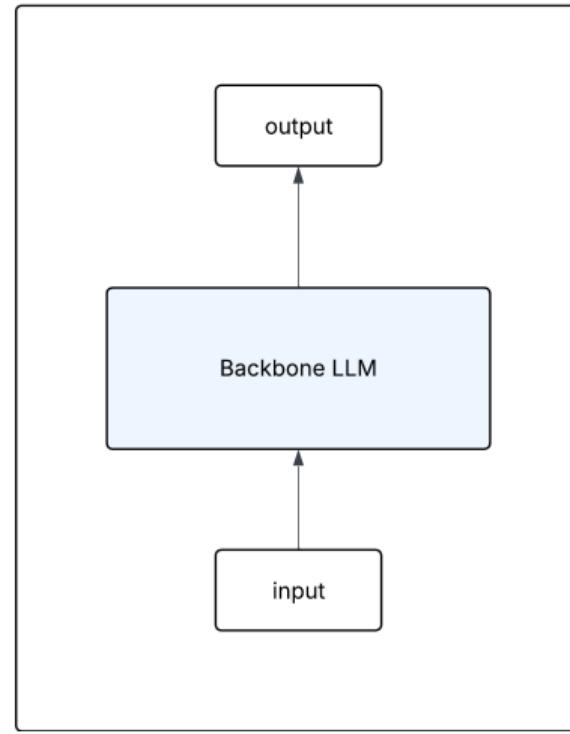
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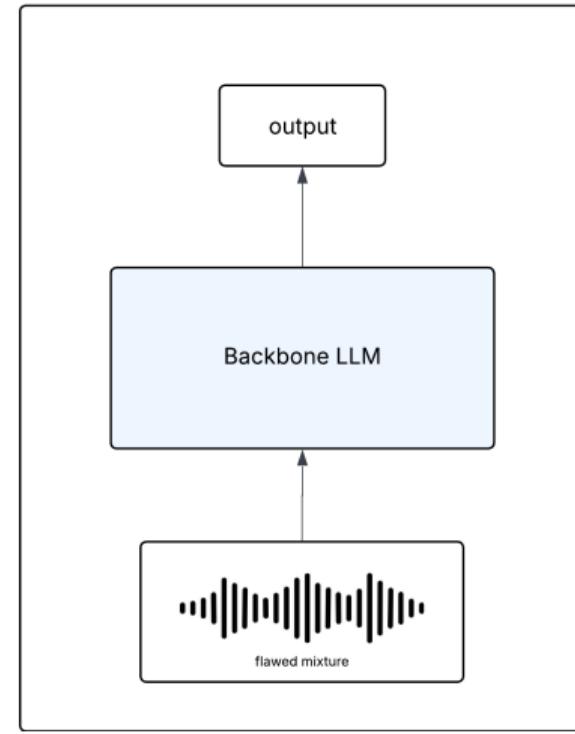
- **LLM as backbone:** A pretrained LLM such as Qwen2 [16] as the backbone.
- **Input:** A Flawed mix.
- **Output:** A structured response containing advice pointing out the flaws and suggesting solutions.
- **Architecture:** Cascade approach, with the LLM as the backbone.
- **Training strategy:** Supervised fine-tuning using PEFT (Parameter-Efficient Fine-Tuning), specifically LoRA [17] or QLoRA [18].



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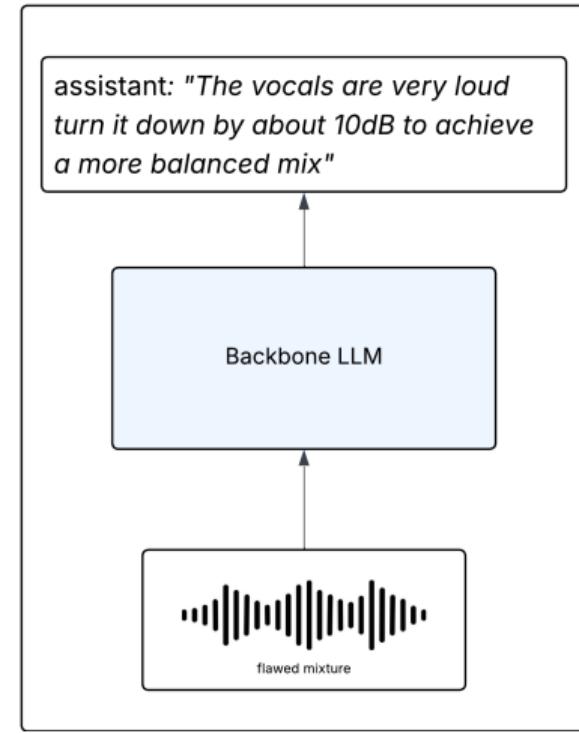
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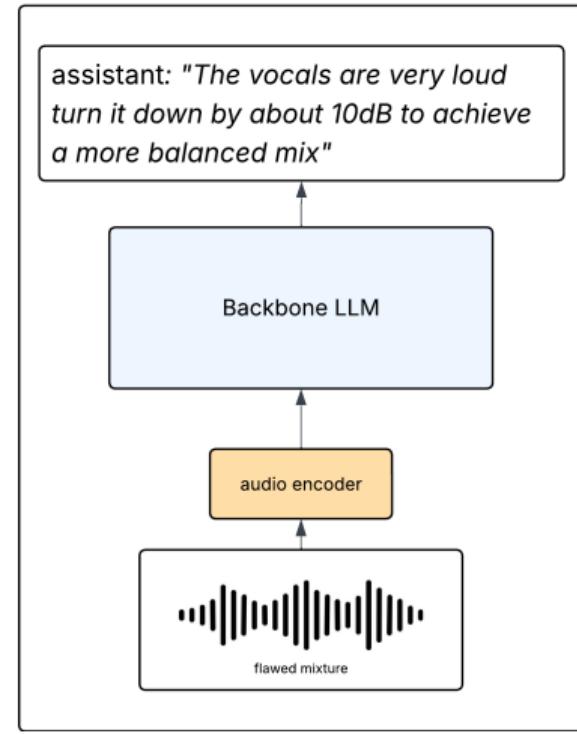
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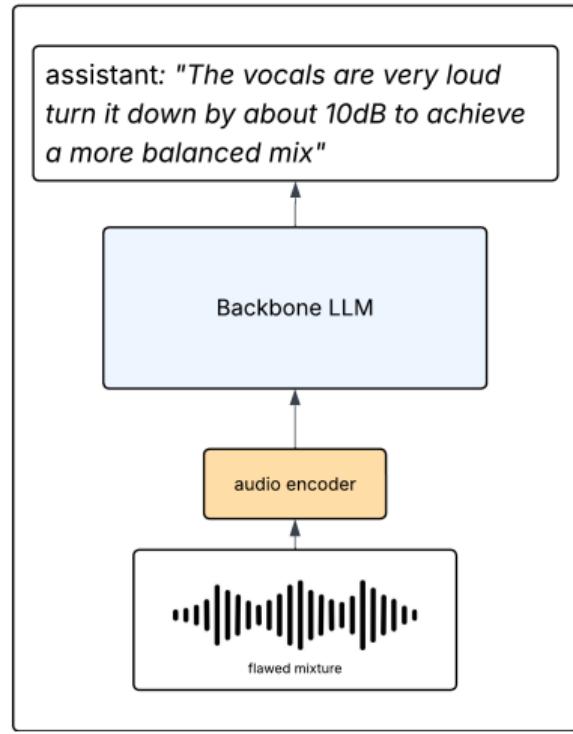
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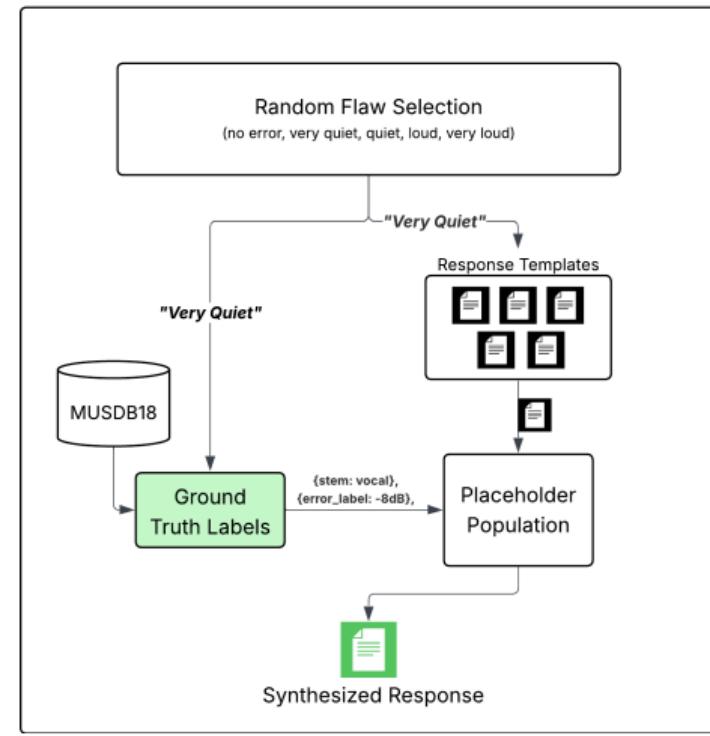


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Dataset Synthesis (The Response)



- **Overview:** Our training data is synthetically generated by formulating diverse input-response pairs.
 - **Flaw-Driven Templating:** A “Flaw Category” is randomly selected, dictating the structure and content of the response.
 - **Dynamic Population:** Template variables such as **stem names** and **suggested gain values** are automatically populated.

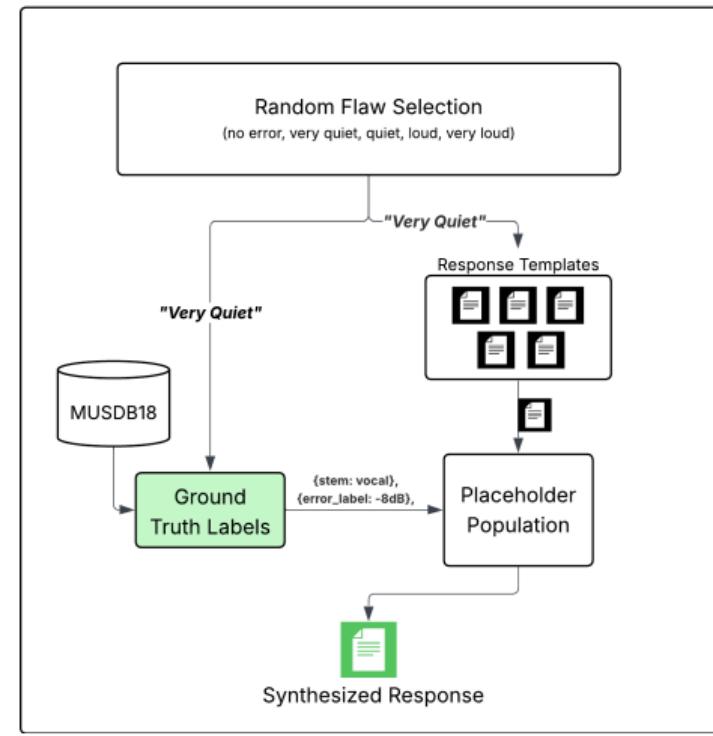


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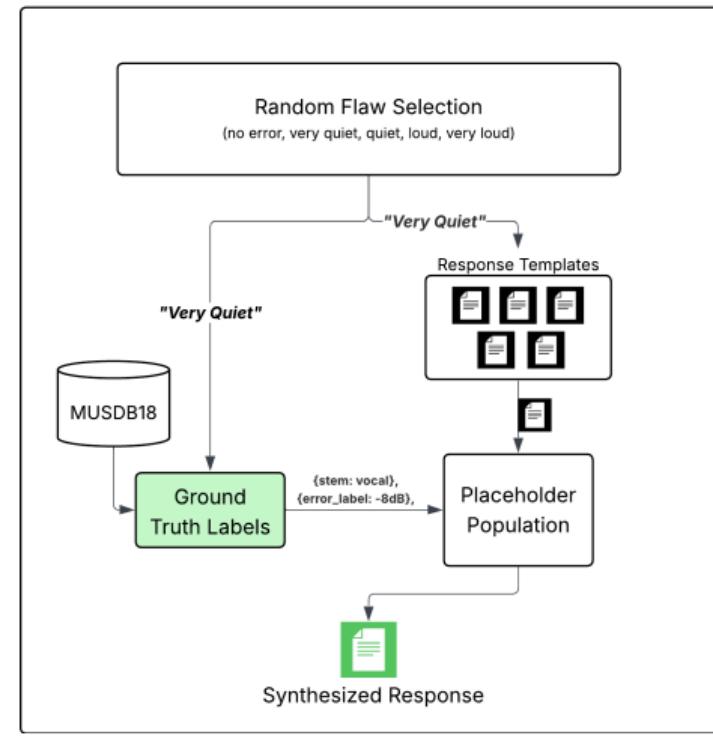


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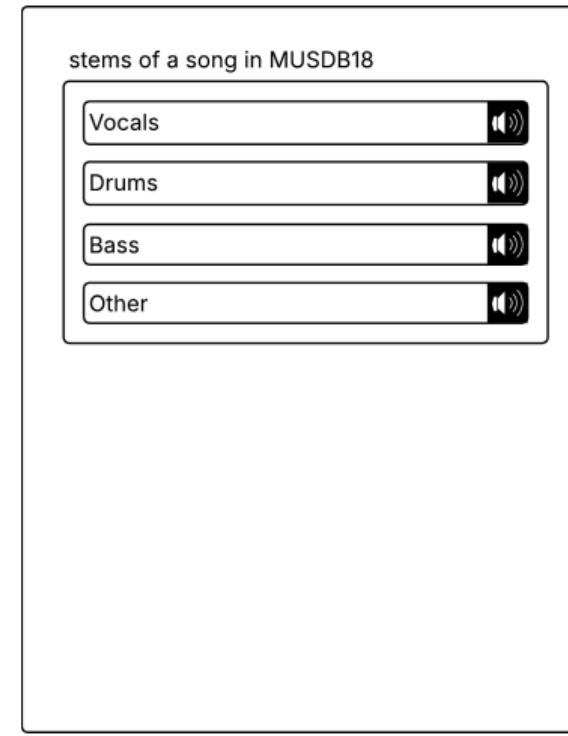
Key Mixing Flaw Categories for Synthesis:

- **No Error:** *“The mix sounds balanced.”*
- **Quiet:** *“The vocal is too quiet.”*
- **Very Quiet:** *“The vocal is much too quiet.”*
- **Loud:** *“The bass is too loud.”*
- **Very Loud:** *“The bass is much too loud.”*

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Dataset Synthesis (The Flawed Mix)

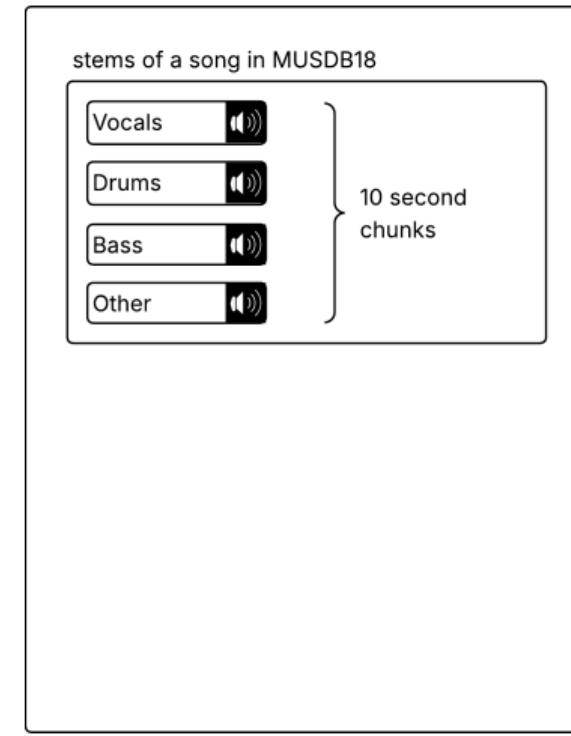
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- Chunk a song into 10-second segments.
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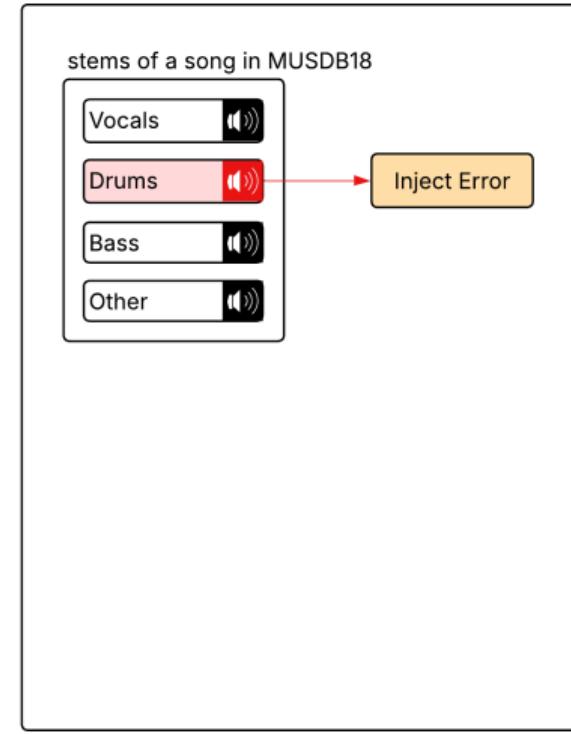
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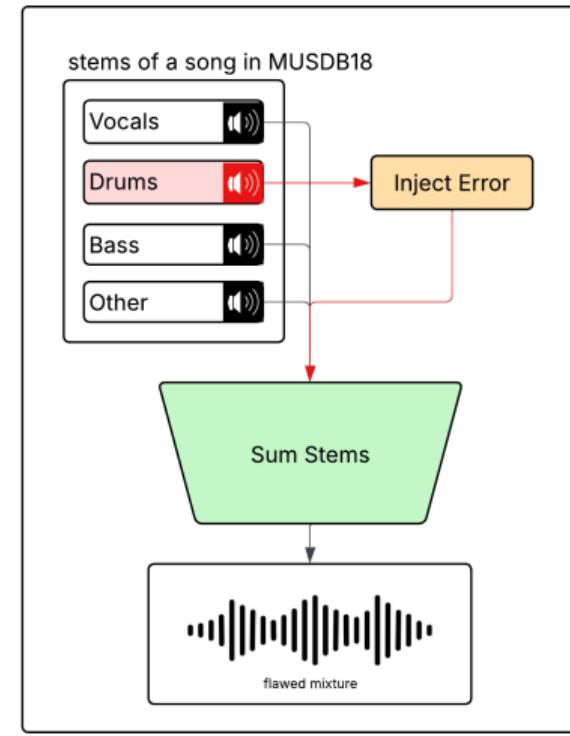
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Human Evaluation

■ **Participants:** Semi-professional audio engineers and producers

■ **Evaluation Criteria:**

- Effectiveness: How well does the advice address the mixing challenge?
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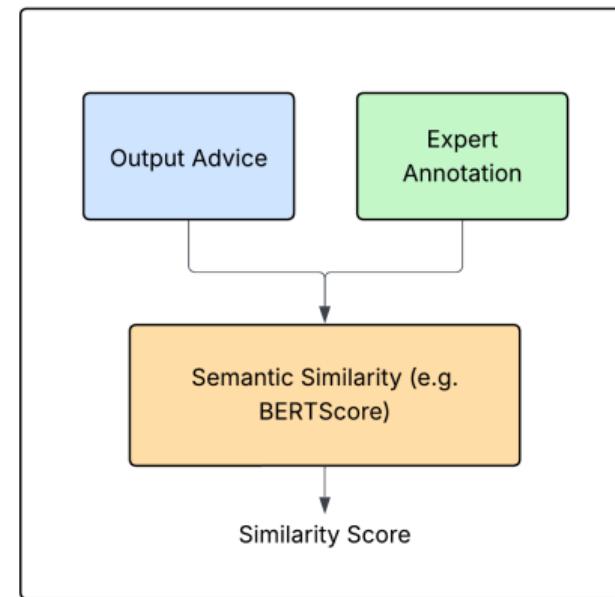
Compares generated advice to expert annotations for relevance.

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Uses a frontier LLM (e.g., Gemini-2.5, GPT-4o) to rate advice quality using a predefined rubric.

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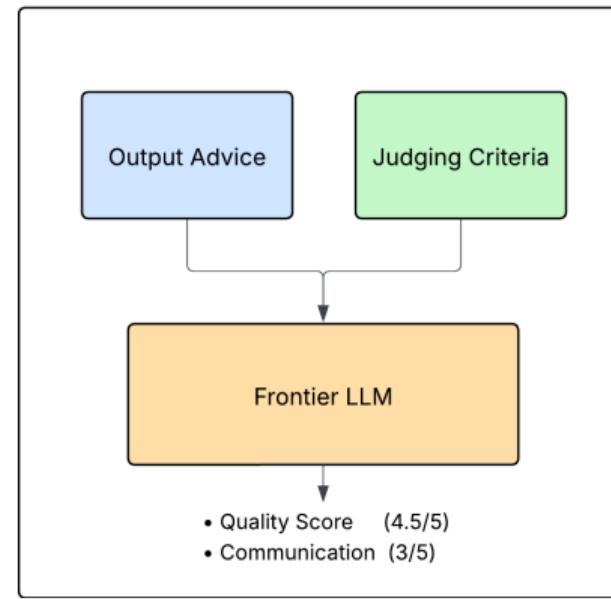
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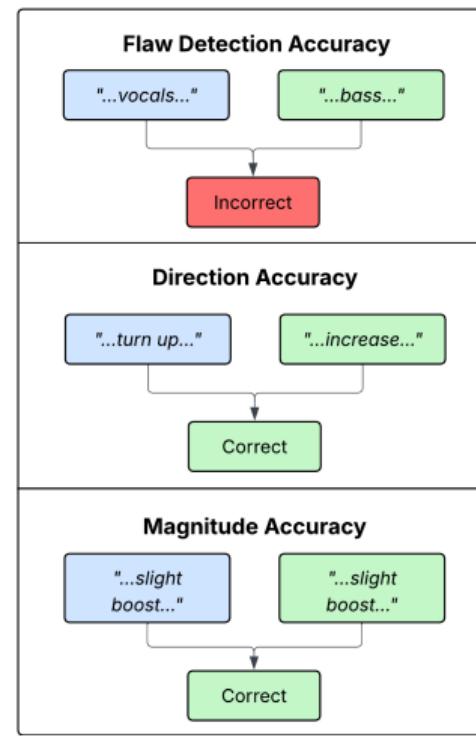
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- **Advisory, Not Prescriptive:** Evaluation focuses on the usefulness of the textual advice, not the numeric accuracy of specific gain predictions.
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- ~~Initial codebase and data loading pipeline setup.~~
- ~~Partial fine-tuning pilot experiments and architecture testing.~~
- Finalize architecture and execute all remaining experiments.
- Submit and obtain IRB approval for human studies.
- Conduct the human evaluation study with audio professionals.

Next Semester

- Submit to a conference, web interface, and Hugging Face deployment.
- Establish better ground-truth dataset that is more representative of professional mixes.
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Questions

Thank you!

Questions?