

# MixingBuddy: A Multimodal LLM for Mix Critique and Advice

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Fall 2025 Project Proposal



Georgia Tech · College of Design

Center for  
Music Technology

# Brief Introduction

## overview

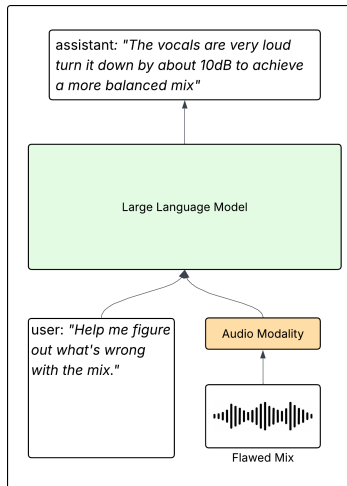
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- This research develops a **multimodal** system that leverages a **pre-trained LLM** to analyze **raw audio** and provide actionable feedback on **flawed mixes**.
- As a starting point, we focus on generating advice for **gain-balancing** only, with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



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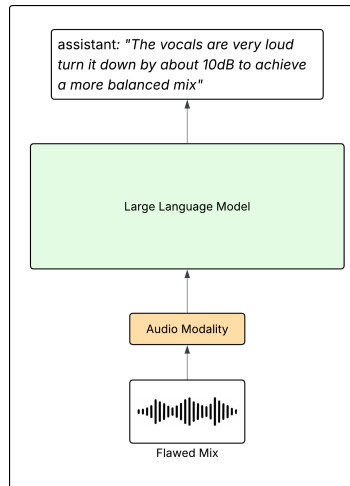
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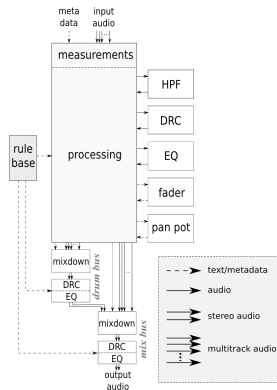
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### Rule-Based and Traditional Machine Learning Systems

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

### Deep Learning Architectures

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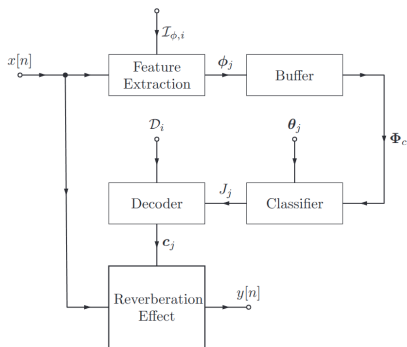


Fig. 1. Reverb application.

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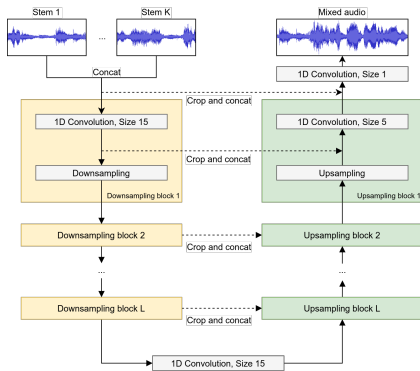
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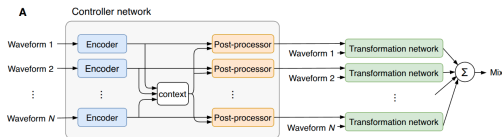
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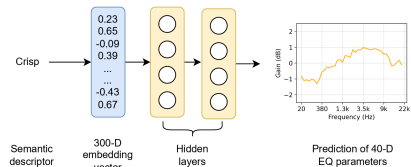


# Automatic Mixing Review

## 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]

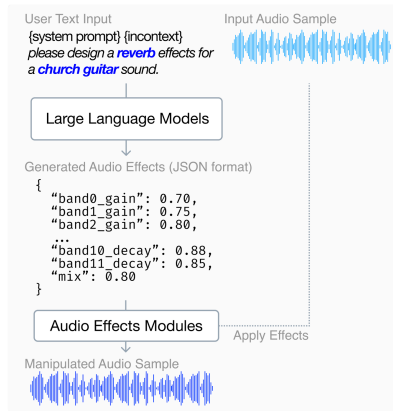


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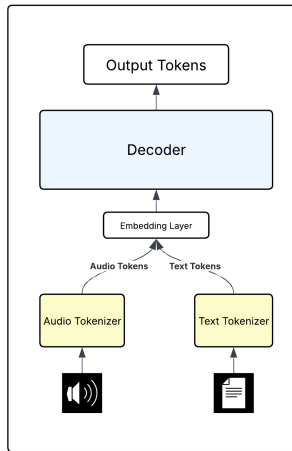
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## Architectural Approaches for Audio-Language Models

- 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].

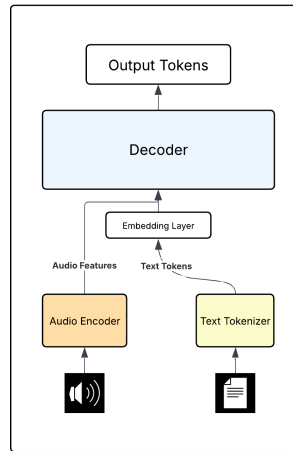


Unified Approach

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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"?
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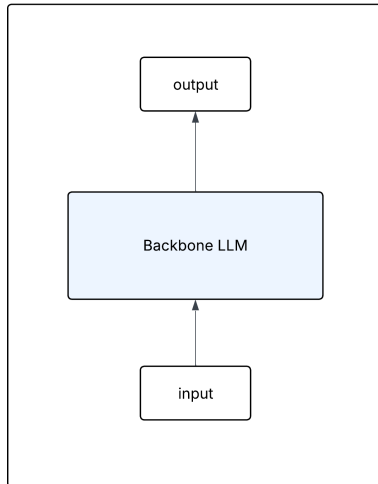
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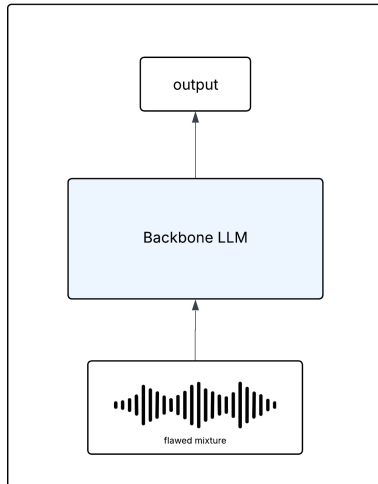
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- **Input:** A Flawed mix.
- **Output:** A structured response containing advice pointing out the flaws and suggesting solutions.
- **Architecture:** Cascade approach, with the LLM as the backbone.
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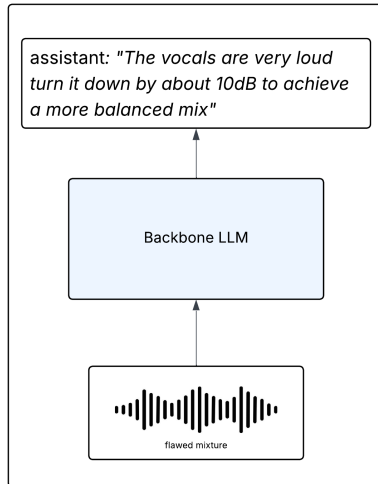
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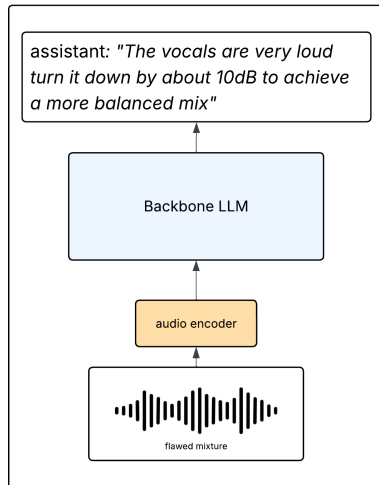
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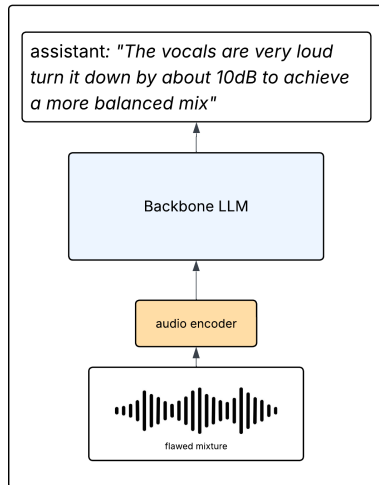
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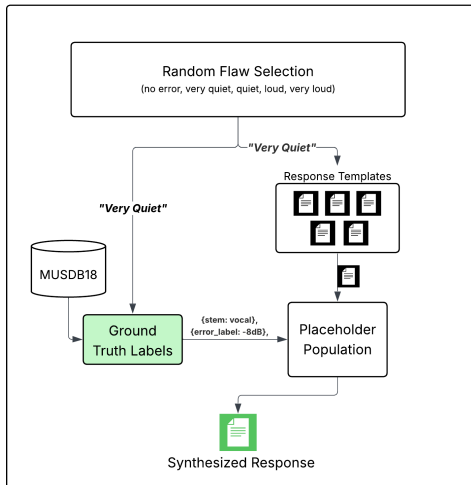
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## Dataset Synthesis (The Response)

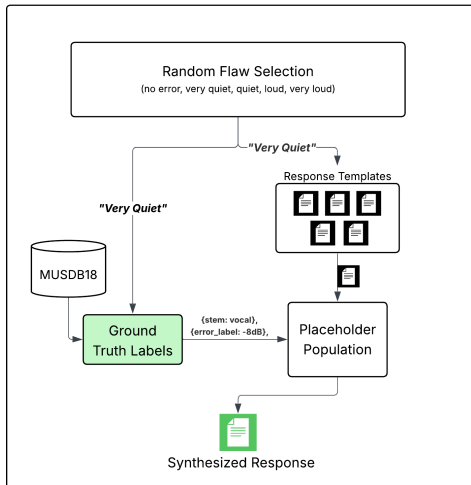
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- Flaw-Driven Templating:** A “Flaw Category” is randomly selected, dictating the structure and content of the response.
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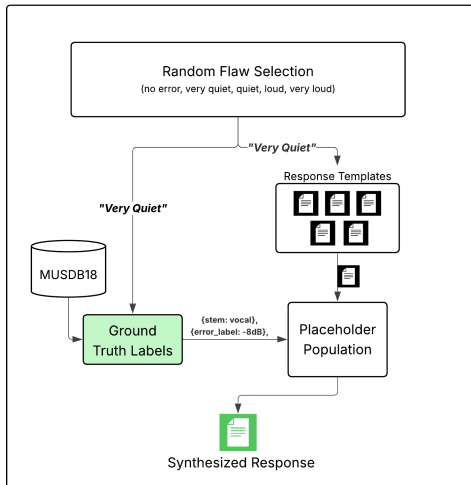




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### Key Mixing Flaw Categories for Synthesis:

- **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.”*

# Proposed Method

## Dataset Synthesis (The Flawed Mix)

- **Dataset:** A multitrack dataset like MUSDB18 [19].
- Chunk a song into 10-second segments.
- Inject an error of  $\pm n$  dB on a non-anchor track based on Flaw Categories.
- Sum the stems to get the flawed mix.

stems of a song in MUSDB18

Vocals



Drums



Bass



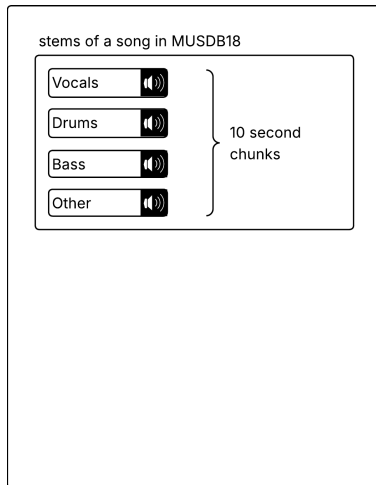
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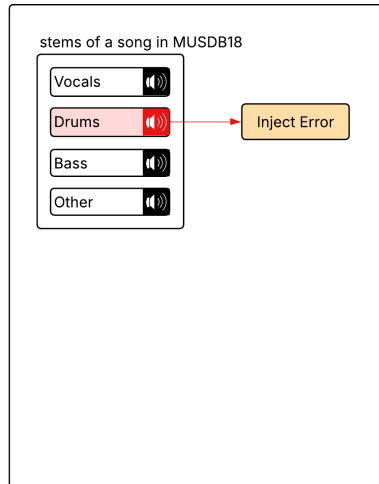
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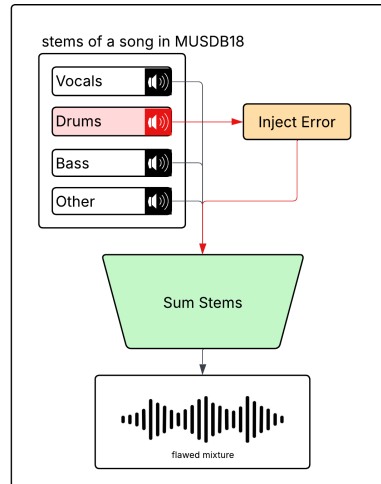
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## Human Evaluation

- **Participants:** Semi-professional audio engineers and producers
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  - **Effectiveness:** How well does the advice address the mixing challenge?
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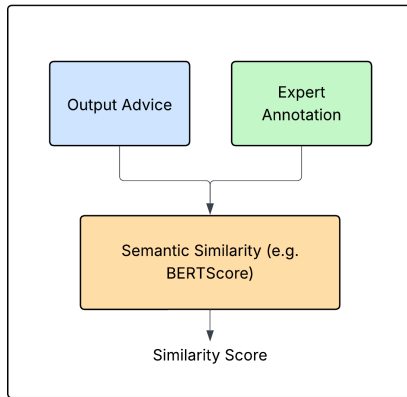
Compares generated advice to expert annotations for relevance.

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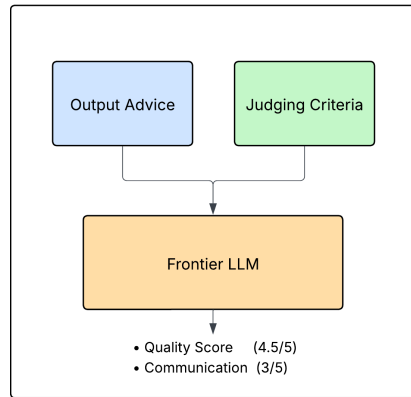
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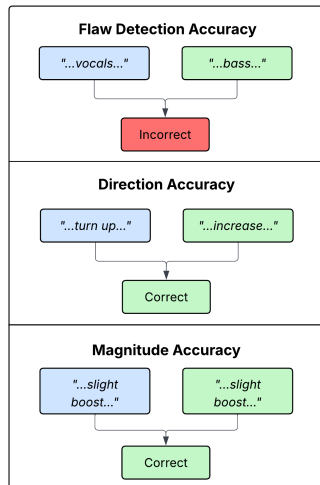
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- **Advisory, Not Prescriptive:** Evaluation focuses on the usefulness of the textual advice, not the numeric accuracy of specific gain predictions.
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- ~~Initial codebase and data loading pipeline 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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- [2] E. Chourdakis and J. Reiss, "A machine-learning approach to application of intelligent artificial reverberation," en, *Journal of the Audio Engineering Society*, vol. 65, no. 1/2, pp. 56–65, Feb. 2017, ISSN: 15494950. DOI: [10.17743/jaes.2016.0069](https://doi.org/10.17743/jaes.2016.0069).
- [3] E. Chourdakis and J. Reiss, "Automatic music signal mixing system based on one-dimensional wave-u-net autoencoders," en, 2022. DOI: [10.1186/s13636-022-00266-3](https://doi.org/10.1186/s13636-022-00266-3). [Online]. Available: [https://www.researchgate.net/publication/366902955\\_Automatic\\_music\\_signal\\_mixing\\_system\\_based\\_on\\_one-dimensional\\_Wave-U-Net\\_autoencoders](https://www.researchgate.net/publication/366902955_Automatic_music_signal_mixing_system_based_on_one-dimensional_Wave-U-Net_autoencoders).
- [4] C. J. Steinmetz, J. Pons, S. Pascual, and J. Serrà, "Automatic multitrack mixing with a differentiable mixing console of neural audio effects," no. arXiv:2010.10291, Oct. 2020, arXiv:2010.10291 [eess]. DOI: [10.48550/arXiv.2010.10291](https://doi.org/10.48550/arXiv.2010.10291). [Online]. Available: <http://arxiv.org/abs/2010.10291>.
- [5] A. Chu, P. O'Reilly, J. Barnett, and B. Pardo, "Text2fx: Harnessing clap embeddings for text-guided audio effects," no. arXiv:2409.18847, Feb. 2025, arXiv:2409.18847 [eess]. DOI: [10.48550/arXiv.2409.18847](https://doi.org/10.48550/arXiv.2409.18847). [Online]. Available: <http://arxiv.org/abs/2409.18847>.
- [6] S. Venkatesh, D. Moffat, and E. R. Miranda, "Word embeddings for automatic equalization in audio mixing," en, *Journal of the Audio Engineering Society*, vol. 70, no. 9, pp. 753–763, Nov. 2022, ISSN: 15494950. DOI: [10.17743/jaes.2022.0047](https://doi.org/10.17743/jaes.2022.0047).
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# references

- [9] J. Melechovsky, A. Mehrish, and D. Herremans, "Sonicmaster: Towards controllable all-in-one music restoration and mastering," no. arXiv:2508.03448, Aug. 2025, arXiv:2508.03448 [eess]. DOI: [10.48550/arXiv.2508.03448](https://arxiv.org/abs/2508.03448). [Online]. Available: <http://arxiv.org/abs/2508.03448>.
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- [11] P. K. Rubenstein et al., "Audiopalm: A large language model that can speak and listen," no. arXiv:2306.12925, Jun. 2023, arXiv:2306.12925 [cs]. DOI: [10.48550/arXiv.2306.12925](https://arxiv.org/abs/2306.12925). [Online]. Available: <http://arxiv.org/abs/2306.12925>.
- [12] Z. Du et al., "Lauragpt: Listen, attend, understand, and regenerate audio with gpt," no. arXiv:2310.04673, Jul. 2024, arXiv:2310.04673 [cs]. DOI: [10.48550/arXiv.2310.04673](https://arxiv.org/abs/2310.04673). [Online]. Available: <http://arxiv.org/abs/2310.04673>.
- [13] D. Zhang et al., "Speechgpt: Empowering large language models with intrinsic cross-modal conversational abilities," no. arXiv:2305.11000, May 2023, arXiv:2305.11000 [cs]. DOI: [10.48550/arXiv.2305.11000](https://arxiv.org/abs/2305.11000). [Online]. Available: <http://arxiv.org/abs/2305.11000>.
- [14] S. Liu, A. S. Hussain, Q. Wu, C. Sun, and Y. Shan, "M<sup>2</sup>ugen: Multi-modal music understanding and generation with the power of large language models," no. arXiv:2311.11255, Dec. 2024, arXiv:2311.11255 [cs]. DOI: [10.48550/arXiv.2311.11255](https://arxiv.org/abs/2311.11255). [Online]. Available: <http://arxiv.org/abs/2311.11255>.
- [15] Y. Gong, H. Luo, A. H. Liu, L. Karlinsky, and J. Glass, "Listen, think, and understand," no. arXiv:2305.10790, Feb. 2024, arXiv:2305.10790 [eess]. DOI: [10.48550/arXiv.2305.10790](https://arxiv.org/abs/2305.10790). [Online]. Available: <http://arxiv.org/abs/2305.10790>.
- [16] A. Yang et al., *Qwen2 technical report*, 2024. arXiv: [2407.10671](https://arxiv.org/abs/2407.10671) [cs.CL]. [Online]. Available: <https://arxiv.org/abs/2407.10671>.
- [17] E. J. Hu et al., "Lora: Low-rank adaptation of large language models," no. arXiv:2106.09685, Oct. 2021, arXiv:2106.09685 [cs]. DOI: [10.48550/arXiv.2106.09685](https://arxiv.org/abs/2106.09685). [Online]. Available: <http://arxiv.org/abs/2106.09685>.

# references

- [18] T. Dettmers, A. Pagnoni, A. Holtzman, and L. Zettlemoyer, "Qlora: Efficient finetuning of quantized llms," *arXiv preprint arXiv:2305.14314*, 2023, arXiv:2305.14314. [Online]. Available: <https://arxiv.org/abs/2305.14314>.
- [19] Z. Rafii, A. Liutkus, F.-R. Stöter, S. I. Mimilakis, and R. Bittner, *Musdb18-hq - an uncompressed version of musdb18*, Aug. 2019. DOI: [10.5281/zenodo.3338373](https://doi.org/10.5281/zenodo.3338373). [Online]. Available: <https://doi.org/10.5281/zenodo.3338373>.

# Questions

Thank you!

Questions?