

# Project Deliverable D1.3

## Team

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## Repository

<https://github.com/poinka/NewWave.git>

## Project Topic

Natural Language Music Recommender System

## What has been done so far

- Performed exploratory data analysis on Jamendo (large-scale, ~1 TB with many songs) and Sound Descriptor (tiny, higher-quality) to decide dataset roles; used Jamendo for training and some evaluation batches, and Sound Descriptor only for evaluation to gauge generalization under description shifts.
- Implemented a parquet-by-parquet pseudo-streaming pipeline to overcome storage limits and unstable direct streaming, enabling per-file load-and-delete to simulate Hugging Face streaming during training with controllable batches.
- Extended the CLAP context window via a linear schedule: audio clips increased from 10 seconds toward 240 seconds and text descriptions expanded beyond 77 tokens, approximately adding one second per several batches.
- Ran long-context CLAP training on A100 for about one day, covering roughly 10% of the dataset; observed both stagnation and sharp validation gains across runs on Jamendo subsets and Sound Descriptor, and continued training to improve generalization.
- Iterated the lyrics/description retrieval encoder: removed Longformer due to poor convergence and inefficiency; trained all-MiniLM-L6-v2 with hard negatives (artists, then tags); tested an average-pooled text variant; switched to bge-m3, increased batch for stability, and extended to 25 epochs for best results.

# Results

## CLAP

- Explored two datasets with distinct roles: Jamendo as a very large corpus (~1 TB with many songs) used for training and some evaluation batches, and Sound Descriptor as a tiny but higher-quality set used only for evaluation to assess generalization under description style shifts.
- Due to storage limits and unstable direct streaming from the hub, implemented a parquet-by-parquet pseudo-streaming loader: load each parquet shard, process batches, then delete to free space, effectively simulating streaming with controllable batch flow on both laptops and InnoDataHub servers.
- With this pipeline, a full training day on A100 covered roughly 10% of the dataset, confirming throughput feasibility while indicating more epochs/passes are needed for stable generalization.
- Training objective was to extend CLAP beyond its pretraining limits of 10 s audio and 77-token text by applying a linear schedule that gradually increases audio duration toward 240 s and expands text length, roughly adding about one second of audio per several batches while similarly loosening the token cap on descriptions.
- Validation behavior varied by run: some runs showed stagnation while others produced sharp gains on Jamendo subsets and on Sound Descriptor evaluation, consistent with sensitivity to schedule, batch regime, and data stream composition; extended training is ongoing to consolidate gains and improve generalization across description styles.

Metrics:

MRR: 0.1122

Recall@k:

Recall@1: 0.0442

Recall@5: 0.1521

Recall@10: 0.2487

Recall@20: 0.3820

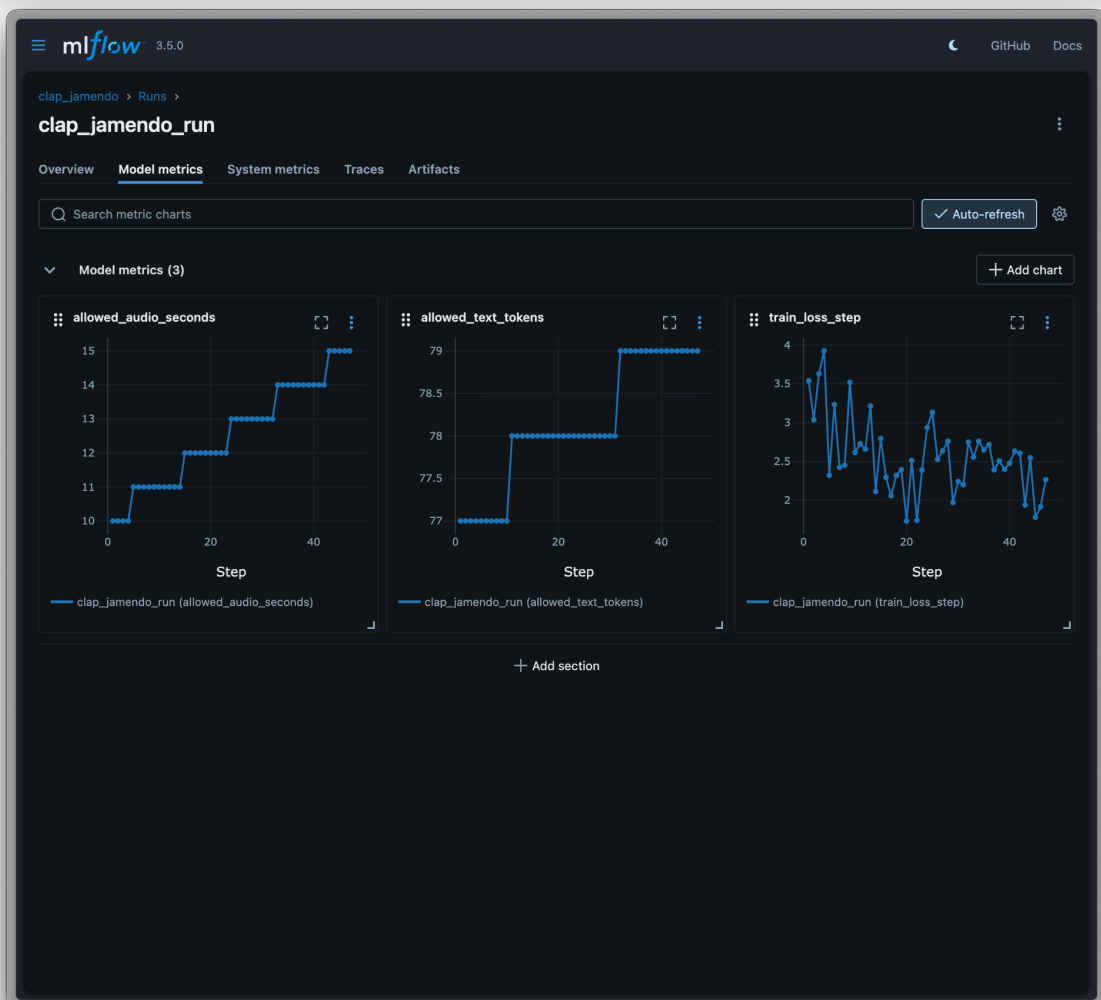
Precision@k:

Precision@1: 0.0442

Precision@5: 0.0304

Precision@10: 0.0249

Precision@20: 0.0191



## SOTA comparison

- Against recent state of the art on the same evaluation set (Song Describer), CLaMP 3 (2025) reports MRR = 0.1985; our best current retrieval run reaches MRR = 0.1169 on the same dataset, i.e., about 1.76× behind the latest SOTA; the paper also reports a CLAP fine-tuned baseline of 1.131 (same family as ours), which aligns with our setup and supports the fairness of the comparison (source: <https://arxiv.org/pdf/2502.10362v1> )
- Relative to prior CLAP results, our R@10 = 0.2487 is within 4.4% of the CLAP 2023 figure (0.2601), and our R@1 = 0.0442 matches CLAP 2023 exactly, indicating competitive early-rank retrieval while long-context training continues (source: <https://arxiv.org/pdf/2311.10057> )

## Lyrics/Description retrieval

### Model iterations and rationale

- Removed the Longformer model because it failed to train, was very slow, and consumed too much memory; as a result, lyric embeddings had to be refreshed only every 2000 steps, which was inefficient. Literature checks also showed it is not well-suited for semantic retrieval unless further adapted on very large datasets.
- Switched to all-MiniLM-L6-v2. Training was significantly faster, and it was possible to add hard negatives (first by artists, then by tags) without excessive memory pressure. With 256-token truncation for both descriptions and lyrics (the best-performing length), the best test results were:

MRR: 0.0685  
Recall@1: 0.0288  
Precision@1: 0.0288  
Recall@5: 0.0984  
Precision@5: 0.0197  
Recall@10: 0.1479  
Precision@10: 0.0148  
Recall@20: 0.2065  
Precision@20: 0.0103

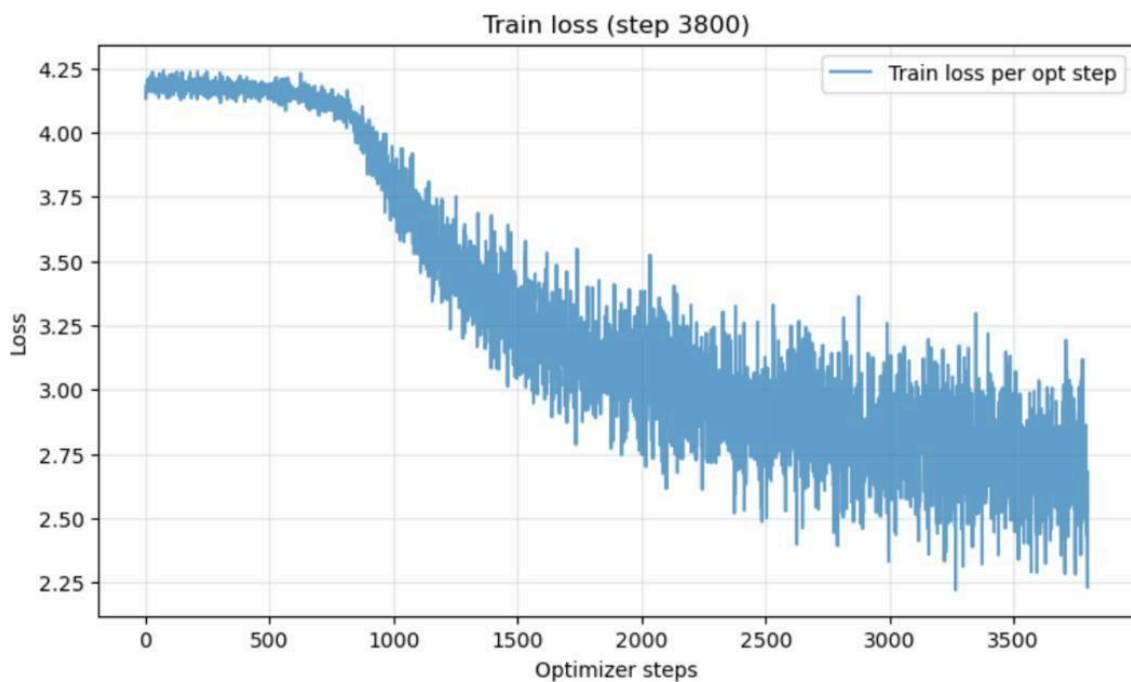
The problem was that we cut the text, description, and lyrics to 256 tokens because the model produced the best results at that length, which meant a lot of the text simply didn't make it into the model.

- Tried average pooling over the full text; training became much longer and did not pay off:

MRR: 0.0086  
Recall@1: 0.0016  
Precision@1: 0.0016  
Recall@5: 0.0084  
Precision@5: 0.0017  
Recall@10: 0.0164  
Precision@10: 0.0016  
Recall@20: 0.0298  
Precision@20: 0.0015

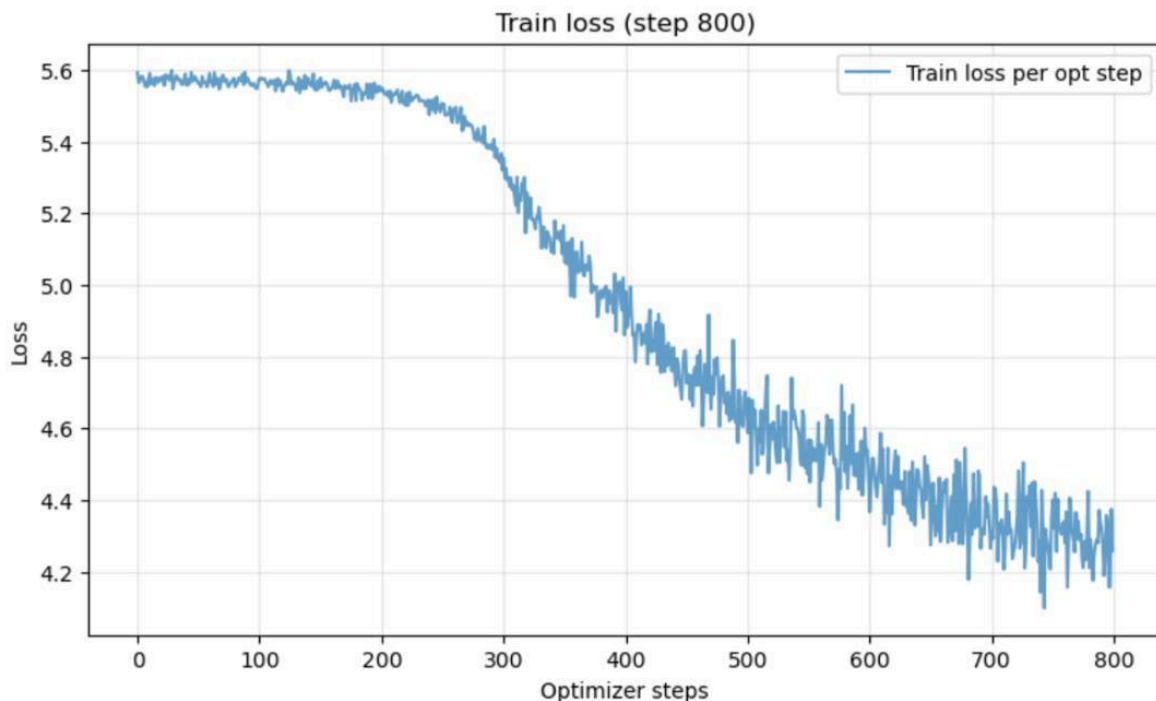
- Moved to a larger model that can take both the full description and lyrics: bge-m3. Trained separate heads for lyrics and descriptions. Initial results:

MRR: 0.0871  
Recall@1: 0.0404  
Precision@1: 0.0404  
Recall@5: 0.1258  
Precision@5: 0.0252  
Recall@10: 0.1849  
Precision@10: 0.0185  
Recall@20: 0.2499  
Precision@20: 0.0125



- Increased batch size to stabilize training; metrics improved:

MRR: 0.0932  
 Recall@1: 0.0441  
 Precision@1: 0.0441  
 Recall@5: 0.1325  
 Precision@5: 0.0265  
 Recall@10: 0.1947  
 Precision@10: 0.0195  
 Recall@20: 0.2641  
 Precision@20: 0.0132

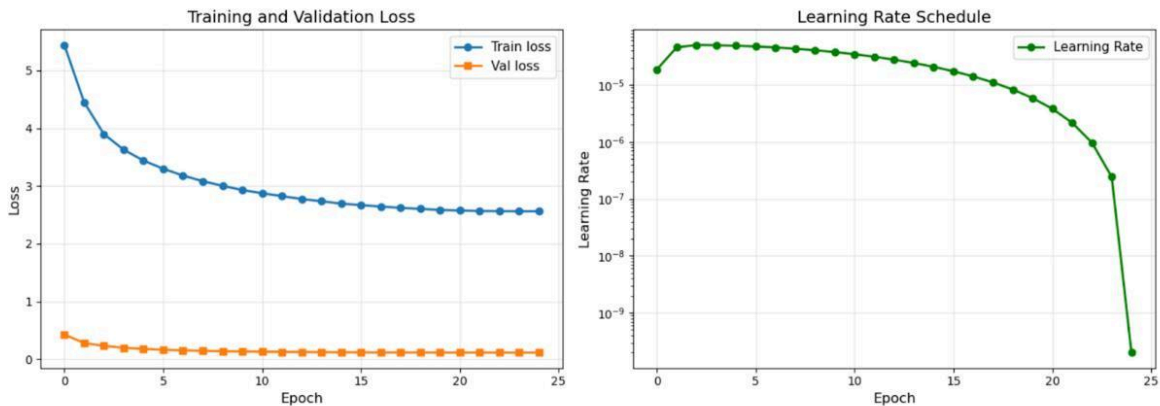


Training: 83% | 102507/124102 [1:38:50<18:43, 19.22it/s, loss=4.2588, step=800, lr=4.94e-05]

- Extended training to 25 epochs (from 5 previously), achieving the best results so far:

MRR: 0.1169  
 Recall@1: 0.0542  
 Precision@1: 0.0542  
 Recall@5: 0.1734  
 Precision@5: 0.0347  
 Recall@10: 0.2496  
 Precision@10: 0.0250  
 Recall@20: 0.3335  
 Precision@20: 0.0167

## last model training



## Work distribution

- **Polina Korobeinikova:** Conduct lyrics/description encoder experiments; removed Longformer; trained all-MiniLM-L6-v2 with hard negatives; tested average pooling; switched to bge-m3, stabilized with larger batch, and scaled to 25 epochs to reach the best metrics.
- **Janna Ivanova:** Conduct EDA on Jamendo and Sound Descriptor; determined dataset roles (Jamendo for training and partial eval, Sound Descriptor for eval) to test generalization under description shifts. Create this report.
- **Vladislav Kalinichenko:** Built the parquet-based pseudo-streaming loader to handle storage/streaming constraints; orchestrated long-context CLAP runs and monitored validation behavior and throughput on A100.

## Plan for the Next Weeks

- develop a demo application
- download songs for a retrieval
- connect endpoints to work with the application
- train a retrieval system that combines two current models