

# AlterDub – Conceptual Training Data Flow

This document provides a clean, architecture-level understanding of how audio data flows from raw WAV files to training-ready tensors in AlterDub, focusing on Modules 2, 3, and 4. It is intended as a visual and mental reference.

## 1. Raw Audio to Clips

Long recordings are first segmented into manageable audio clips, typically between 3 to 8 seconds. Each clip represents one training example candidate.

## 2. Framing (Time Axis Creation)

Each audio clip is divided into overlapping frames using a fixed hop size (e.g., 10 ms). This step introduces the time axis (T). The total number of frames T depends on clip duration and hop size.

## 3. Feature Extraction (Module 2)

For every frame in a clip, two primary features are computed:

- **Mel Spectrogram:** A time-frequency representation capturing phonetic content, formants, timbre, and articulation. Stored as a 2D structure [T, Mel\_Bins].
- **Pitch (F0):** One fundamental frequency value per frame, capturing intonation and prosody. Stored as a 1D structure [T].

Each clip produces one mel file and one pitch file, both already containing all frames for that clip.

## 4. Dataset Indexing (Module 3)

Each clip is assigned a speaker ID. Metadata records link together mel path, pitch path, speaker ID, and frame length (T). At this point, one clip equals one sample.

## 5. VCDataset (Module 4)

VCDataset defines what one training sample is. It loads all features for a single clip and returns mel, pitch, speaker ID, and length as structured numerical containers.

## 6. Batching & T\_max (Collate Function)

Training uses batches (groups of samples). Because clips have different lengths, the collate function pads samples to match the longest sequence in the batch, called T\_max.

- Batch size (B): Number of samples grouped together (e.g., B = 4).
- T\_max: Maximum frame length among samples in the batch.
- Mask: Identifies real frames versus padded frames.

The resulting tensors have shapes like: mel [B, T\_max, 80], pitch [B, T\_max], speaker\_id [B].

## 7. Model Training

The model consumes these batched tensors. Masks ensure that learning happens only on real frames, not padding. This standardized tensor flow enables efficient GPU training and stable convergence.

**Key Mental Model:** One audio clip equals one sample. A batch is multiple samples stacked along a batch dimension. T\_max exists to make variable-length speech trainable.