

# Training Day 7 Report

## Speech To Text Generation:

### Purpose

This Python script captures a 5-second audio recording, sends it to AssemblyAI (a speech-to-text API service), transcribes the audio into text, and saves the result in a file named output.txt.

It involves:

- **Recording audio** using your microphone
- **Uploading** that audio to AssemblyAI
- **Polling** (repeated checking) the API until the transcription is complete
- **Saving** and displaying the final transcribed text
- **Technologies Used**

Module	Purpose
sounddevice	: Captures audio from the microphone
soundfile	: Saves the recorded audio to a .wav file
requests	: Sends HTTP requests to AssemblyAI's REST API
time	: Adds delay between polling attempts
AssemblyAI	: Performs speech-to-text transcription using cloud-based AI

### Why Use This Approach?

- Local recording gives user control over data capture.
- AssemblyAI offers highly accurate AI-powered transcription.
- The script runs entirely in terminal, making it lightweight and beginner-friendly.

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### Input

- No user input is needed — the script records audio for 5 seconds automatically.

### Output

- `recorded.wav`: A saved recording of the user's voice.
- `output.txt`: A file containing the transcribed text of the spoken audio.

## Step-by-Step Explanation of the Code

### Step 1: Record Audio

python  
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**duration**: How many seconds to record (5 seconds here).

**fs (sampling rate)**: 44,100 samples per second (CD-quality audio).

**filename**: Name of the file where audio will be saved.

- **sd.rec(...)**: Starts recording audio into a NumPy array.
- **sd.wait()**: Waits until the recording is finished.
- **sf.write(...)**: Saves the array into a .wav file.
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**WAV Format** is used because AssemblyAI expects standard formats like .wav

### Step 2: Upload Audio to AssemblyAI

Prepares the API key and endpoint to upload the audio file.

- Sends the audio file to AssemblyAI's /upload endpoint.
- Gets a secure `audio_url` from the response, which will be used for transcription.

### Step 3: Start Transcription Request

- Sends a transcription request to the AssemblyAI API.
- The request includes the `audio_url` of the uploaded file.
- API returns a unique `transcript_id` used to track the progress.

## Polling:

Polling is a **core strategy** when dealing with async APIs that need processing time. It:

- Enables your script to wait smartly.
- Gives you control over pacing (with sleep).
- Is safer and more predictable than using callbacks in APIs that don't support them.

## Polling and Saving Transcription:

- This continuously checks (polls) the status of the transcription.
- AssemblyAI usually takes a few seconds to process the request.
- When status is "completed", the text result is extracted and written to output.txt.
- The transcription is printed to the terminal as well.
- Prints the error if transcription failed. Waits for 3 seconds before polling again (to avoid overloading the API).

## Why is Polling Used in This Script?

AssemblyAI **doesn't immediately return** the transcription. Instead, it:

1. Accepts the audio for transcription.
2. Starts processing it.
3. Asks the client to poll a specific **/transcript/{id}** endpoint to know when it's done.

# Theory Behind the Components & Future Enhancements:

## 1. Audio Recording

- Capturing analog sound via microphone and converting it into a digital signal using a sample rate.
  - Audio is stored in a .wav file in PCM format.
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## 2. REST API

- A REST API allows a client (Python script) to communicate with a server (AssemblyAI) using standard HTTP methods:
    - POST to send data
    - GET to retrieve status
  - JSON format is used for sending/receiving structured data.
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## 3. Speech-to-Text (STT)

- AssemblyAI uses deep learning models (likely transformers or CNN-RNN hybrids) trained on thousands of hours of audio to:
  - Detect spoken words
  - Convert them into accurate transcriptions