Documentation - Django Intern Evaluation, applyin.co - Anant Yadav

1 Performance Metrics

• Average Transcription Latency: 0.84 seconds

• Average Accuracy: 0.89

2 Limitations of Product

- 1. Scalability of the webpage is not checked.
- 2. How the webpage responds to multiple users concurrently is not tested.
- 3. Implementing groups for real-life usage.
- 4. Challenge: Making it work on Vercel, a static website deployer, and Web-Socket deployment challenges.

3 Challenges and Learnings

- 1. Django Realtime Communication:
 - Technologies: Django Channels, ASGI, Consumer, WebSocket.
- 2. Realtime Transcription with Deepgram API:
 - Technology: Deepgram API integration for real-time transcription, accuracy, and latency calculation.
- 3. Deploying Django on Vercel:
 - Approach: Vercel Django Implementation.

4 ChatGPT/LLM Integration Documentation

4.1 Application Overview

- **Prompt:** Describe the Real-Time Audio Transcription Webpage application.
- Output: Provided an overview of the application's functionality, including the record button, Whisper API integration, and user interface design.

4.2 Recorder

• **Prompt:** Implement views for recording, transcribing, and displaying results.

• Output:

```
# Real-Time Audio Transcription Webpage
       ## Setup JavaScript for Audio Recording and Whisper
           → API Integration
       1. Open the HTML file where you want to implement
           \hookrightarrow audio recording and Whisper API integration
          (e.g., transcripti..../home.html).
       2. Add the following script to handle audio

→ recording using Recorder.js.

          Make sure to adjust the code based on the
              \hookrightarrow Whisper API documentation.
       '''html
       <!-- (e.g., transcripti..../home.html). -->
       <!-- ... (previous HTML code) ... -->
13
14
       <script>
            // Implement JavaScript for audio recording and
               → Whisper API integration here
            // Adjust the code based on the Whisper API
               \hookrightarrow documentation
18
            // Example: Basic audio recording using
               → Recorder.js
            var recorder;
            navigator.mediaDevices.getUserMedia({ audio:
               → true })
                .then(function (stream) {
23
                    recorder = new Recorder(stream);
24
                })
```

```
.catch(function (err) {
26
                     console.log('Error accessing
27
                         → microphone:', err);
                 });
28
            function startRecording() {
30
                 recorder && recorder.record();
31
32
33
            function stopRecording() {
34
                 recorder && recorder.stop();
                 // Handle the audio data and send it to the
                     \hookrightarrow server for transcription
            }
37
        </script>
38
39
```

4.3 Error Handling

- **Prompt:** How can robust error handling be implemented for poor audio quality or network interruptions?
- Output: Suggested using try...except blocks, providing clear user feedback for poor audio quality or network interruptions.

4.4 Performance Metrics

- **Prompt:** What key performance metrics should be measured and reported, and how can they be calculated?
- Output: Advised measuring transcription latency and accuracy, with suggestions on using timestamps for latency and evaluating text with Levenshtein distance for accuracy.

4.5 Closing WebSocket Connection

- **Prompt:** How to automatically close the WebSocket connection after 90 seconds and restart it when the user clicks the record button?
- Output: Proposed using a timer or setTimeout function to close the connection after 90 seconds, with the initiation of a new connection on the record button click.

4.6 Handling Rate Limit Error

• **Prompt:** How to handle the 'RateLimitError' from the OpenAI Whisper API?

• Output: Advised implementing try...except blocks to catch the 'Rate-LimitError' and providing user feedback to check the plan and billing details.

4.7 Django Channels WebSocket Connection

- **Prompt:** How to configure WebSocket connection in Django Channels consumer?
- Output: Shared a general structure for connecting to Deepgram using Django Channels, including connection, disconnection, and receiving functions.

4.8 Connecting to Deepgram with Django Channels

- Prompt: How to connect to Deepgram using Django Channels and handle WebSocket events?
- Output: Presented a code snippet for connecting to Deepgram, handling CLOSE and TRANSCRIPT-RECEIVED events, and raising exceptions for connection issues.

4.9 Code Generation

• Task: Write Python code to send real-time audio transcriptions using Deepgram API through Django Channels and WebSockets. Include handling for transcript and accuracy.

• Subtasks:

- 1. Connect to Deepgram API using provided credentials.
- 2. Implement WebSocket consumer to handle real-time transcription events.
- 3. Extract transcript and accuracy from received data and send them through WebSocket.
- 4. Ensure proper connection handling and message reception.

4.10 Transcription Latency Calculation

• Task: Provide a Python function to calculate transcription latency based on Deepgram API duration field.

• Subtasks:

- 1. Extract duration field from the transcription result.
- 2. Return the duration as the transcription latency in seconds.

5 Useful Links:

- Django Channels
- \bullet Live Transcript using Deepgram API
- $\bullet \ \ {\bf Real time \hbox{-} Audio \hbox{-} Transcriber}$
- $\bullet\,$ Deployable Vercel link
- $\bullet~$ Video Link
- Whisper Real time