Project Brief: Al Speech-to-Text App (Including Voice Commands)

Project Title: AI Speech-to-Text App (Including Voice Commands)

Project Overview:

The AI Speech-to-Text App is a dynamic voice assistant—style application designed to enable real-time transcription, summarisation, and interactive voice-based control. Users will be able to engage in spoken conversations with the app, transcribe live meetings, livestreams, and recorded video/audio content (up to 1 hour), and receive AI-generated summaries of what was discussed. The app will also include a custom voice command system and optional voice training module that improves recognition for individual users, creating a smarter and more personal interaction experience.

Objectives:

- 1. Develop a voice-interactive application that transcribes real-time and pre-recorded audio up to 60 minutes in length.
- 2. Enable natural conversational interactions, allowing users to speak directly to the app and receive spoken or textual responses.
- 3. Implement voice command recognition for performing specific in-app functions through speech.
- 4. Integrate AI-generated summaries that concisely reflect the contents of transcribed audio or conversations.
- 5. Offer a voice training feature that allows users to record samples and improve recognition of their individual voice profiles.
- 6. Ensure a responsive and accessible user interface for uploading content and viewing live interactions or summary outputs.

Key Features:

- 1. **Two-Way Voice Assistant Functionality:** Supports natural conversations between the user and AI via microphone input.
- 2. **Transcription Engine:** Converts audio from meetings, livestreams, or uploaded recordings into accurate text.
- 3. **Al Summarisation:** Uses large language models (e.g. Gemini or Whisper + summarisation layer) to deliver **contextual summaries** of conversations.
- 4. **Voice Training Module:** Allows users to upload or record voice samples to improve speech recognition accuracy over time.
- 5. **Multi-Modal Input:** Accepts **microphone input, file uploads**, or **live audio streams** as sources.
- 6. **Copy and Export Options:** Transcripts and summaries can be copied to clipboard or downloaded in .txt or .pdf format.

Technical Specifications:

- 1. Programming Language: Python
- 2. Frameworks and Libraries:
 - Flask/Django: Web application framework.
 - o SpeechRecognition, PyAudio, whisperx or Vosk for STT (speech-to-text)
 - o webrtcvad for voice activity detection
 - o NLP libraries (e.g., SpaCy or NLTK): for language understanding and intent recognition.
- 3. Database: SQLlite for storing user-defined configurations and logs.
- 4. **Deployment:** Docker containerization and cloud service deployment (e.g., AWS or Azure).
- 5. **Version Control:** Git for source code management.

Expected Outcomes:

- 1. A fully functional AI-powered voice assistant and transcription tool with real-time command capabilities.
- 2. Enhanced speech recognition accuracy through user-specific voice training.
- 3. Streamlined meeting and media analysis via automatic transcription and smart summarisation.
- 4. A responsive UI offering live feedback, searchable transcripts, and multi-format exports.

Risks and Mitigations:

- 1. **Recognition Errors:** Include fallback text input and voice re-activation prompts; use robust models with adaptive training.
- 2. Latency or Lag: Optimise streaming and model inference for real-time responsiveness.
- 3. **Privacy & Security:** Encrypt all user voice data and provide clear options to manage and delete stored recordings.
- 4. **Audio Input Challenges:** Support a wide range of file formats and stream configurations with input validation and cleanup handling.