# **Hackathon Project Phases**

## **Project Title:**

Audio transcriber app using open ai whisper

## **Team Name:**

Team curious crew

## **Team Members:**

- Bodgam.Rishika
- Radapaka.Sushmitha
- Emmadi.Tejaswi

## **Phase-1: Brainstorming & Ideation**

## **Objective:**

Develop an AI powered tool to transcribe an audio into text and summarize it.

## **Key Points:**

#### 1. Problem Statement:

Large corporations struggle document team meetings, client calls.

#### 2. Proposed Solution:

- An Al-powered application using open ai whisper to provide key takeaways from meeting calls.
- Accurately transcribes multi accent discussions.

#### 3. Target Users:

- Businesses and professionals.
- Content creators.
- Government and public services.

#### 4. Expected Outcome:

 A functional Al powered transcriptor app to generate accurate, well formatted text from spoken audio.

## **Phase-2: Requirement Analysis**

### **Objective:**

- 1.Develop an automated system that converts spoken language from an audio file into text.
- 2.Detects the language of the transcribed text.
- 3.summarize the translated text using Open AI GPT.

### **Key Points:**

#### 1. Technical Requirements:

- Programming Language: Python.
- o Backend: python,ffmpeg,torch,whisper,langid,transformers.
- Frontend: Stream lit Web Framework
- Database: Not required initially.

#### 2. Functional Requirements:

- Accepts an audio file(e.g.,MP3,WAV etc..)
- o Detects the spoken language.
- o Displays the original transcription.
- Summarize the text.

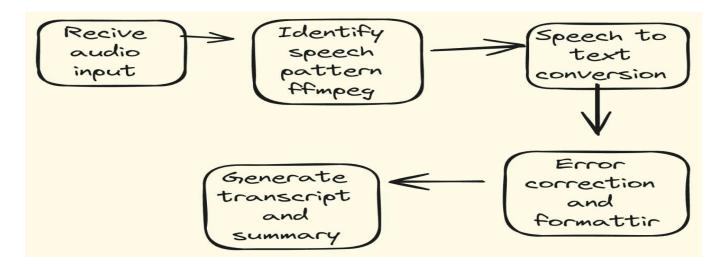
#### 3. Constraints & Challenges:

- Internet requirements.
- Providing a smooth UI experience with Streamlit.
- OpenAl GPT\_4 API charges based on usage
- Processing large audio files.

## **Phase-3: Project Design**

## **Objective:**

Develop the architecture and user flow of the application.



## **Key Points:**

#### 1. System Architecture:

- User uploads an audio file
- Query is processed using open ai whisper
- o Al model fetches and processes the data.
- The frontend displays transcript, detected language, summary

#### 2. User Flow:

- Step 1: User upload an audio file
- Step 2: The backend transcripts speech to text
- Step 3: Some other models help finding the errors in sentences and fix them
- Step 4: Result gives the transcript and key takeaways from given audiofile

#### 3. UI/UX Considerations:

Minimalist, user-friendly interface for seamless navigation.

## **Phase-4: Project Planning (Agile Methodologies)**

## **Objective:**

Break down development tasks for efficient completion.

Sprint	Task	Priority	Duration	Deadline	Assigned To	Dependencies	Expected Outcome
Sprint 1	Environment Setup	High	6 hours (Day 1)	End of Day 1	Sushmitha Tejaswi	Python, Streamlit setup	established & working
Sprint 1	Frontend UI Development	 Medium	2 hours (Day 1)	End of Day 1	Rishika	response format finalized	Basic UI with input fields
Sprint 2	Audio file upload	High	3 hours (Day 2)	Mid-Day 2	Tejaswi Rishika	UI elements ready	Search functionality with filters
Sprint 2	Error Handling & Debugging	High	5 hours (Day 2)	End of the day 2	Members 3	UI inputs	Improved stability
Sprint 3	Testing & UI Enhancements	 Medium	2 hours (Day 2)	Mid-Day 2	Tejaswi Sushmitha	UI layout completed	Responsive UI, better user experience
Sprint 3	Final Presentation & Deployment	Low	1 hour (Day 2)	End of Day 2	Entire Team	Working prototype	Demo-ready project

### **Sprint Planning with Priorities**

### **Sprint 1 – Setup & Integration (Day 1)**

- ( High Priority) Set up the environment & install dependencies.
- ( Medium Priority) Build a basic UI with input fields.

## Sprint 2 – Core Features & Debugging (Day 2)

- ( High Priority) Implement search of audio file uploading.
- ( High Priority) Debug API issues & handle errors in queries.

## Sprint 3 – Testing, Enhancements & Submission (Day 2)

- ( Medium Priority) refine UI, & fix UI bugs.
- ( Low Priority) Final demo preparation & deployment.

## **Phase-5: Project Development**

## **Objective:**

Implement core features of the Audio transcriber App.

## **Key Points:**

#### 1. Technology Stack Used:

o Frontend: Streamlit

Backend: ffmpeg,python,torch.Programming Language: Python

#### 2. Development Process:

Setup ffmpeg/transformers/langid.

o Implement key features : speech recognition ,language detection & translation.

Summarization.

#### 3. Challenges & Fixes:

Challenge: low quality or noisy audio formats.

**Fix:** use noise reduction techniques before transcription.

o Challenge: large audio files may cause performance bottlenecks.

**Fix:** limit file size and implement chunk-based processing for long recordings.

## **Phase-6: Functional & Performance Testing**

## **Objective:**

Ensure that the AutoSage App works as expected.

Test Case ID	Category	Test Scenario	Expected Outcome	Status	Tester
TC-001	Functional Testing	A 4 seconds audio file(EN) is uploaded	To Give correct transcription	✓ Passed	Rishika
TC-002	Functional Testing	A 10 seconds audio file(EN) of different accent is uploaded	To give correct transcription and summary	✓ Passed	Sushmit ha
TC-003	Performance Testing	Large audio data files	To give response quickly		Tejaswi
TC-004	Bug Fixes & Improvements	A 10 seconds audio file(mix lang) is uploaded	To give correct transcription with summary	⚠Needs optimization	Sushmit ha

TC-005	Final Validation	Ensure UI is responsive across devices.	UI should work on mobile & desktop.	<b>✓</b> Passed	Tejaswi
TC-006	Deployment Testing	Host the app using Streamlit Sharing	App should be accessible online.	Deployed	Entire team

## **Final Submission**

- 1. Project Report Based on the templates
- 2. Demo Video (3-5 Minutes)
- 3. GitHub/Code Repository Link
- 4. Presentation