Voice-Activated Al Assistant Using OpenAl and Whisper

A Real-Time Voice Interaction System
Ranishree Anegundi

Overview

Objective:

- Create a voice-activated assistant capable of:
- Speech recognition
- Natural language processing
- Voice-based response generation

Key Features:

- Real-time audio transcription
- Wake-word detection
- Integration with OpenAI's ChatGPT
- Text-to-speech response system

System Architecture

Components:

- 1. Audio Recording: Captures user audio input.
- 2. Speech Recognition: Transcribes audio using Whisper model.
- 3. Wake-Word Detection: Detects specific trigger words to activate the assistant.
- 4. NLP Processing: Sends the transcribed query to ChatGPT.
- 5. Text-to-Speech (TTS): Converts the AI response to speech.

Flow Diagram:

 User speaks → Audio Recording → Whisper Transcription → Wake Word Detection → ChatGPT Query → TTS Response → User hears response.

Tools and Technologies

Programming Language: Python

- Libraries and Frameworks:
- torch, numpy: Data handling and processing.
- speech_recognition: Audio input handling.
- whisper: OpenAI's speech-to-text model.
- gTTS and pydub: Text-to-speech and audio playback.
- openai: Interact with ChatGPT.

Hardware Requirements:

- Microphone
- GPU (optional for Whisper model optimization)
- - Environment:
- .env file to securely load OpenAI API keys.

Code Walkthrough - Main Modules

Audio Recording:

- Captures audio continuously.
- Uses speech_recognition for microphone input.

Transcription and Wake Word Detection:

- Transcribes speech using Whisper.
- Detects wake word before processing queries.

NLP with ChatGPT:

- Sends the query to OpenAI's GPT-3.5-Turbo model.
- Receives a natural language response.

Text-to-Speech Response:

- Uses gTTS for speech synthesis.
- Plays the response audio using pydub.

Code Highlights

```
Wake Word Detection:
     if predicted_text.lower().startswith(wake_word.lower()):
       # Strip wake word and send query
ChatGPT Integration:
     payload = {
       'model': 'gpt-3.5-turbo',
       'messages': [{'role': 'user', 'content': prompt}],
       'max tokens': 300,
       'temperature': 0.7,
     response = requests.post('https://api.openai.com/v1/chat/completions', headers=headers,
    ison=payload)
    - Speech Synthesis:
     mp3 obj = gTTS(text=answer, lang='en', slow=False)
     mp3 obj.save('reply.mp3')
     play(AudioSegment.from mp3('reply.mp3'))
```

Demo Workflow

Step 1: Start the program.

Step 2: Say the wake word (e.g., 'Hey

Computer').

Step 3: Ask a question or give a command.

Step 4: Wait for the AI to respond with synthesized speech.

Challenges and Solutions

Challenges:

- Real-time audio processing.
- API communication delays.
- TTS performance and accuracy.

Solutions:

- Efficient threading for parallel tasks.
- Optimized Whisper model for quick transcription.
- Use of gTTS and pydub for seamless audio playback.

Future Enhancements

Improvements:

- Multilingual support.
- Advanced wake-word models using machine learning.
- Enhanced TTS quality with neural-based models.
- Integration with IoT devices for smart home automation.

Conclusion

Summary:

- Successfully implemented a voice-activated assistant.
- Combines cutting-edge speech recognition and NLP.
- Practical application in smart assistants and automation.

Questions?