

Voice FAQ Chatbot – Case Study Report

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An in-depth explanation of the architecture, libraries, and workflow for voice-enabled Streamlit chatbot.

Project Overview

This app is a Streamlit-based voice-enabled chatbot that:

- Accepts voice or text input.
- Transcribes voice to text (STT).
- Retrieves an answer from a PDF-based FAQ using semantic search.
- Speaks the answer aloud (TTS).
- Maintains a transcript of the conversation.

Core Python Files

1. app.py – main Streamlit app
2. answer_engine.py – gets answers from FAQ
3. faq_processor.py – loads PDF and performs question matching
4. speech_utils.py – handles speech-to-text and text-to-speech
5. requirements.txt – all dependencies listed

Libraries & Purpose

Library	Why It's Used
streamlit	To build the web app interface
streamlit-webrtc	For real-time audio capture using WebRTC
numpy	For audio array manipulation
soundfile	To save NumPy audio to `.wav` for STT
speechrecognition	To convert speech (audio) to text using Google
pyttsx3	Offline TTS engine (speaks text)
gtts	Google online TTS (returns MP3/base64)
pydub	To play audio files inside Python
PyMuPDF	To read and extract text from PDF FAQ
sentence-transformers	Semantic embedding for advanced Q&A

	(future-proof)
scikit-learn	For TfidfVectorizer and cosine similarity

Module Breakdown

app.py – Main Interface Logic

Responsibilities:

- Microphone input via streamlit-webrtc
- Voice-to-text transcription
- Calls ask_ai() with query (voice or text)
- Plays both user's question and AI's answer aloud
- Displays chat + transcript

Key Functions:

- AudioProcessor: Collects raw audio frames into a NumPy array
- ask_ai(): The central function that
 - Speaks user's input
 - Queries the AnswerEngine
 - Speaks the answer
 - Updates UI and transcript
- st.chat_input: Accepts fallback typed input

answer_engine.py – Query Router

Responsibilities:

- Initializes a FAQProcessor
- Provides the best matched answer to a question
- Handles exceptions and fallback responses

faq_processor.py – PDF-Based QA Model

Responsibilities:

- Extracts Q&A pairs from a PDF
- Uses TF-IDF vectorization + cosine similarity to find the closest match
- Returns matched answer if similarity is above a threshold

Why TF-IDF?

Efficient and fast for short text similarity tasks like FAQs. Ideal for lightweight models.



speech_utils.py – Voice Tools

Functions:








- tts_play(text): Uses pyttsx3 to speak text directly (offline)
- stt_transcribe_numpy(audio, rate): Saves NumPy audio → WAV → Google STT → text

- `tts_play_to_bytes(text)`: Uses gTTS to generate MP3 in memory as base64

Workflow Overview

User →  Speak or  Type
→ [STT via Google if voice]
→ Question → AnswerEngine
→ FAQProcessor → PDF Search (TF-IDF)
→ Answer → Speak (TTS)
→ Show in UI + Transcript

Notable Features

-  Offline voice output using pyttsx3
-  Online speech recognition using Google STT API
-  Fallback text input
-  PDF-driven FAQ model with semantic text matching
-  Voice output for both user and bot
-  Transcript logging
-  WebRTC-based mic capture

 Considerations for Deployment

| Problem | Solution |

|-----|-----|

| Streamlit Cloud does not support full WebRTC | Use fallback via `st_audiorec` or deploy on Render/EC2 |

| STT depends on internet (Google) | Can add Whisper offline STT later |

| TTS sometimes slow with gTTS | Use offline TTS (`pyttsx3`) for instant feedback |

Summary

Your app is a complete voice-enabled AI chatbot, combining:

- Real-time mic input (via WebRTC)
- STT using Google
- TTS via both `pyttsx3` and gTTS
- PDF-based Q&A search via TF-IDF

- Streamlit UI with typed fallback and chat history

Link: <https://voice-ai-chat-box-dbjh7yrn98yvut6pl2dv7p.streamlit.app/>