

About Me - Karim Lalani

• **Home**: Leander, TX

Work: Software Solutions Architect

Background: Full Stack Engineer, Gen Al

 FOSS: LangChain contributor, FastRTC client, Tool Calling wrappers for LLMs

Using LangChain:

Delivered multiple solutions built with Langchain

Socials:

- Linkedin https://www.linkedin.com/in/-karim-lalani/
- Github https://github.com/lalanikarim/
- Medium https://medium.com/@klcoder

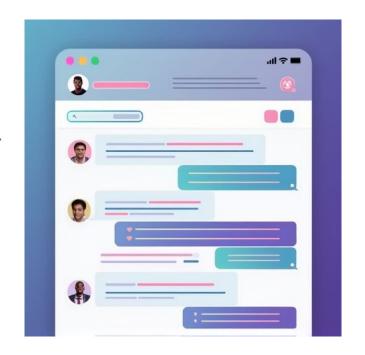




Text-Based Chatbots vs. Voice Agents

Text-Based Chatbots

- **Interaction**: Keyboard input, structured dialogue.
- Accessibility: Limited for visually impaired users.
- Speed: Fast for simple, direct queries.
- Use Cases: FAQs, quick answers, transactional tasks.



Text-Based Chatbots vs. Voice Agents

Voice Agents

- Interaction: Natural speech, hands-free, real-time.
- Accessibility: Enhanced for diverse users (e.g., visually impaired).
- Speed: Dynamic, immersive for complex or multitasking scenarios.
- Use Cases: Multitasking, accessibility, conversational tasks.



OpenAl Voice Mode

- Natural Voice Conversations: Real-time, low-latency interactions for intuitive, human-like dialogue.
- Accessibility & Convenience: Enables
 multitasking and supports users with visual
 impairments or voice-input preferences.
- Tech-Driven Integration: Leverages APIs (e.g., OpenAI Realtime Audio, Deepgram) for speech-to-text, TTS, and sentiment analysis.



Flow of Voice Agents

1. User Speaks

The user initiates interaction by speaking into a microphone.

2. Streaming Audio Data

Audio is captured in real-time and sent to the voice agent system.

3. Voice Activity Detection (VAD)

Detects when the user is speaking (vs. silence or background noise).

4. Speech-to-Text (STT)

o Converts the audio stream into text for processing (e.g., using OpenAl's Realtime API or third-party services).

5. **Processing**

The text is analyzed using NLP or Al models (e.g., ChatGPT) to generate a response.

6. Text-to-Speech (TTS)

The generated response is converted back into audio.

7. Generated Speech Audio Streamed Back

• The audio response is sent to the user in real-time, completing the interaction.

Flow of Voice Agents

Key Notes:

- Real-time streaming ensures low latency for natural conversation.
- VAD and STT are critical for accurate speech recognition and filtering.
- TTS quality impacts the user experience (e.g., naturalness, clarity).

Introducing FastRTC

FastRTC is a platform or framework focused on **real-time communication (RTC)**, built on **WebRTC** (Web Real-Time Communication) and **WebSockets** technology. It enables developers to create **low-latency**, **audio/video streaming and data transfer** for applications like voice agents, video conferencing, live chats, and collaborative tools.

FastRTC simplifies the complexities of WebRTC, making it easier for developers to build real-time, interactive applications with minimal infrastructure overhead. It's ideal for projects requiring high-performance, low-latency communication.

FastRTC

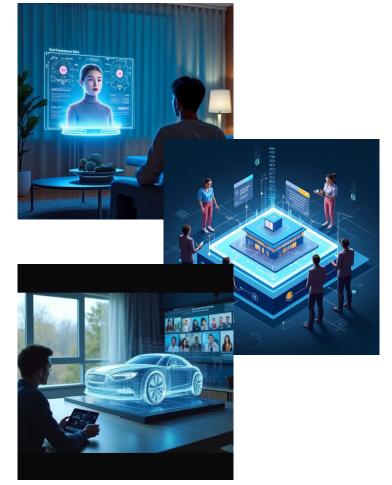
Key Features:

- Real-time audio/video streaming: Facilitates seamless, low-latency communication between users.
- Scalability: Supports large-scale applications with efficient resource management.
- Cross-platform compatibility: Works across browsers, mobile, and desktop environments.

FastRTC

Use Cases:

- Voice agents (e.g., integrating voice chat into Al assistants).
- Live video streaming, virtual meetings, and interactive web apps.
- Real-time data sharing (e.g., collaborative editing, gaming).



Demo

Building an Echo Server with FastRTC

Happy Vibing!

