# DAY -1

**🔧 What is uv?**

uv is a next-gen Python package manager from Astral. It’s:

* ⚡️ **Ultra fast** (written in Rust)
* ✅ Compatible with requirements.txt or pyproject.toml
* 🧼 Replaces pip, venv, pip-tools, virtualenv, poetry, etc.
* 💻 Used internally at OpenAI

**📦 What is pyproject.toml?**

pyproject.toml is the **modern standard configuration file** for Python projects.  
Think of it like the package.json of Python.

It defines:

* Your **project metadata** (name, version, description)
* Your **dependencies**
* Your **build system**
* Your **tooling config** (e.g., for uv, black, ruff, pytest, etc.)

Here’s a clean summary of everything we’ve done so far — you can paste this directly into your notes:

**✅ Step 2: Production-Grade Python Environment Setup (with uv and pyproject.toml)**

**1. 📦 Installed uv**

Used uv as our modern, fast dependency and virtualenv manager.

Install command:

curl -Ls https://astral.sh/uv/install.sh | sh

**2. 🧪 Created .venv using uv**

Created a virtual environment **inside the project folder** (production-grade approach):

uv venv

source .venv/bin/activate

**3. 🛠️ Created pyproject.toml**

Defined all runtime and dev dependencies in one clean config file.

**Runtime dependencies:**

* fastapi
* uvicorn[standard]
* sounddevice
* numpy
* pydantic

**Dev tools:**

* black → autoformatter
* ruff → linter/static checker
* pytest → testing framework

[project]

name = "scribe-ai"

version = "0.1.0"

description = "Real-time voice-to-note AI scribe"

readme = "README.md"

requires-python = ">=3.9"

dependencies = [

"fastapi",

"uvicorn[standard]",

"sounddevice",

"numpy",

"pydantic"

]

[project.optional-dependencies]

dev = [

"black",

"ruff",

"pytest"

]

[tool.uv]

virtualenvs.in-project = true

[tool.black]

line-length = 88

[tool.ruff]

line-length = 88

target-version = "py39"

select = ["E", "F", "I"]

[tool.pytest.ini\_options]

addopts = "-ra -q"

testpaths = ["tests"]

**4. ⚙️ tool.\* in pyproject.toml**

These are **config sections** for dev tools:

* tool.black → sets line length for autoformatting
* tool.ruff → sets linter rules for unused imports, formatting, etc.
* tool.pytest → controls pytest behavior
* tool.uv → makes sure .venv/ is local to the project

**5. 📥 Installed all dependencies (main + dev tools)**

uv pip install .[dev]

This installs everything:

* Runtime dependencies from [project.dependencies]
* Dev tools from [project.optional-dependencies].dev

A **WebSocket** is a special communication protocol that allows **real-time, two-way interaction** between a **client** (like a web browser or mobile app) and a **server**.

Unlike normal HTTP requests (which are one-way: client asks, server replies), WebSockets keep the connection **open**, so **both** sides can send data anytime — perfect for **live updates** like speech apps, chat apps, games, or notifications.

🧠 Think of it like a **phone call**, not sending letters:

* HTTP = letter-based communication (slow, one-way)
* WebSocket = live phone call (fast, two-way)

In a **speech app**, you need real-time interaction. Here’s why HTTP isn't enough:

* **HTTP**: Every time you want to send data (like a chunk of audio), you must open a new connection, send it, and close it. That creates **delays** and **overhead**.
* **WebSocket**: You open **one connection**, and then you can stream data **continuously** in both directions — **perfect for sending live audio and getting live transcripts** back without waiting.

💡 For example:

* Your app sends audio chunks as the user speaks (WebSocket → Server).
* Server replies with transcribed text instantly (Server → WebSocket).

This **back-and-forth flow** is what makes speech feel live

Your **FastAPI app will act as a WebSocket server**. That means:

* You define a **special route** like /ws/audio-stream
* But unlike REST routes (@app.get, @app.post), you use @app.websocket
* This route does **not** expect a request → response flow — it stays open

📌 Inside this WebSocket route:

* You’ll **wait for audio chunks** from the client (await websocket.receive\_bytes() or .receive\_text()).
* You **process** each chunk (e.g., speech-to-text using Whisper/OpenAI).
* You **send back transcription** using await websocket.send\_text().

🔹 Step 6: What is WebRTC? (And how is it different from WebSocket?)

WebRTC (Web Real-Time Communication) is a protocol for direct peer-to-peer communication between browsers or devices. It's used for:

🔊 Voice calls (e.g., Google Meet, Zoom)

🎥 Video streaming

📁 File sharing (without server)

🧠 WebRTC is optimized for media (audio/video) and includes:

Echo cancellation

Latency control

NAT traversal (to go through firewalls)

Encryption (by default)

**At the simplest level, a mic is just a hardware device that listens to sound and converts it into a digital signal your computer or microcontroller can understand.**

So we need to understand the hardware setup better and also start coding to use to laptop mike as a starting point

The task is nice since we need a device that will capture audio.

**So we need a speaker phone**

<https://www.amazon.in/Raspberry-Model-LAN-mHDMI-Supply-MicroSD/dp/B0C7N5DJC1?ref_=v_sp_product_dpx&th=1>

so I have ordered this I think this would be cool to play around and learn , so that we do and cook amazing stuff …intersting times ahead

Note – A design choice has been made for now we need to learn rasbeery pie first and then we need to play the device bit of it

But we need to cook the v1 for system a doctor can open the app and listen start processing and send notes to self and patient

# DAY -2

Started with a small recap on design I mean static method and class method were used in one project just saw that , back to work now.

So feature 1 will be something like this

**🩺 Feature 1: Patient Follow-Up on Telegram (Post-Consultation)**

**🎯 Goal:**

Allow patients to **ask doubts or follow-up questions** on Telegram after their consultation, and **get instant, personalized replies** — powered by your system's backend intelligence.

**🧩 System Flow:**

1. **Session Completed by Doctor**
   * The doctor records and completes a session using your app.
   * Your backend transcribes the audio and stores it with patient metadata (e.g., name, phone number, session ID, summary).
2. **Patient Gets Summary**
   * A message is sent to the patient on Telegram via your bot with:
     + The summary of the consultation.
     + Advice, prescriptions, red flags.
     + A note: “You can ask your follow-up questions here anytime.”
3. **Patient Sends a Message**
   * The patient types a question on Telegram, like:
     + “Can I take this medicine after food?”
     + “I still have a sore throat, should I be concerned?”
4. **Your Backend Handles It**
   * Identifies patient using Telegram ID or linked phone number.
   * Retrieves the session summary + prescription + red flags.
   * Feeds all of that + patient’s question into an LLM API.
   * Receives a smart, patient-friendly answer.
5. **Bot Replies Back**
   * Sends the response to the patient on Telegram.
   * Ends with a safety disclaimer (e.g., “This is an AI response. Please call your doctor if symptoms worsen.”)

**🏗️ 1. High-Level System Design Overview**

**🔊 Input**

* Mobile app or web client (using WebRTC or WebSocket) for capturing audio from doctor (or both doctor & patient).
* Audio streamed to backend **in real-time**.

**⚙️ Backend Core (FastAPI)**

* Audio ingestion via **WebSocket**.
* Realtime transcription (e.g. **OpenAI Whisper** or **Deepgram**).
* LLM summary + prescription generation.
* Message delivery via **Telegram API** or WhatsApp Business API.

**☁️ Infrastructure**

* Host on **Azure Cloud** with:
  + Azure App Service or AKS for backend
  + Blob Storage for audio
  + CosmosDB / MongoDB for session data
  + Azure Queue for async processing
  + Redis for temporary memory/cache

**📶 2. Streaming Audio Design**

| **Component** | **Tech** |
| --- | --- |
| Audio Source | Browser/Mobile Mic |
| Stream Protocol | WebSocket (or WebRTC) |
| Format | audio/webm or wav chunks |
| Transcription | OpenAI Whisper API or Deepgram Streaming API |
| Buffering | Circular buffer in Python / Redis |
| Resilience | Auto-reconnect, retries, chunk validation |

**🧠 3. AI/LLM Processing**

1. **Transcription**
   * POST /transcribe auto-triggered after stream ends.
   * Or done in real-time for streaming APIs like Deepgram.
2. **LLM Summary**
   * Context-aware summarization using OpenAI (GPT-4-turbo).
   * Use custom system prompts (e.g., SOAP-style outputs).
3. **Prescription Generator**
   * Separate LLM prompt/chain to generate:
     + Medicines
     + Dosage
     + Follow-ups
     + Investigations

**📦 4. Scalable Architecture (Cloud-native)**

* **App Server**: FastAPI behind **Gunicorn** + **Uvicorn workers**.
* **Message Queue**: Azure Queue or RabbitMQ for:
  + LLM jobs
  + Transcription
  + Messaging jobs
* **Worker Pool**: Celery workers or FastAPI background tasks.
* **Storage**:
  + Audio files → Azure Blob Storage
  + Summaries → MongoDB or CosmosDB
* **Cache**: Redis (for user session / intermediate transcripts)

**📲 5. Messaging + Telegram Integration**

* One-time bot setup per doctor.
* Transcripts sent via Telegram using sendMessage API.
* Patients can reply → webhook hits POST /telegram/receive
* Use RAG or memory DB to answer follow-ups.

**🧪 6. Testing & Monitoring**

* **Logging**: Use structlog or loguru for production logs.
* **Monitoring**: Azure Monitor or Grafana.
* **Rate Limiting & Auth**:
  + Use API Keys / JWT for doctor access.
  + Rate limit audio ingestion endpoints.

**✅ 7. Security**

* Use HTTPS/WSS for all communication.
* Audio data encrypted in transit and at rest.
* OTP-based access for doctors via phone or email.
* Patient data stored in compliance with basic HIPAA/GDPR principles.

**⏳ 8. Scalability Notes**

| **Feature** | **Scalable With** |
| --- | --- |
| Audio | Chunked upload, queue buffering |
| LLM Processing | Async via Celery / Queue |
| Messaging | Rate-limited, async |
| Frontend | SSR React/Next.js or Streamlit-lite |
| Logging | Logstash / Azure Monitor |

**Todo: we need this product on whats app I do not know how get a whats app business number register and see how u can cook it**