Smart Home for the Aged

Final Project Report

Team 37

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1. Project Overview

The "Smart Home for the Aged" system enhances elderly users' safety and comfort through fall detection, voice command recognition, and automated environmental controls. This intelligent home system creates a safer living environment for elderly residents while providing convenience through hands-free operation.

2. Backend Code Functionality

Key Responsibilities:

- Polls the OM2M audio container (/in-name/voice_command/audio_upload) every 4 seconds for new base64-encoded audio recordings
- Reassembles WAV files from multipart Base64 messages (AUDIO_START, AUDIO_CHUNK, AUDIO_END)
- Uses FasterWhisper for speech-to-text transcription and SentenceTransformer for semantic matching against predefined commands
- Applies a similarity threshold (0.2) to choose the best command, attaches confidence and recognized text
- Executes commands by POSTing content instances back to the appropriate OM2M resources (LED, solenoid, fan)
- Tracks processed sessions via hashing to avoid duplicates
- Data Persistence: A separate Python script periodically fetches OM2M content instances (audio, fall, sensor data) and stores them in a local SQLite database

3. Main Node Code Overview

The ESP32 "main node" firmware handles fall detection and voice recording, uploading both to OM2M:

Setup & Connectivity

- Initializes MPU6050 accelerometer and Wi-Fi
- Provides audible success/error tones

Fall Detection

- Reads accelerometer, detects free-fall (< 0.44 g) followed by impact (> 0.05 g)
- Plays alert tone and uploads event data to OM2M

Touch-Triggered Recording

- On capacitive touch, records 3 seconds of 8 kHz audio via MAX9814 microphone
- Applies simple signal processing for improved clarity

WAV Packaging & Upload

- Builds a 44-byte WAV header
- Splits audio into four Base64 chunks and sends to OM2M
- Provides feedback & reliability through tones at key stages
- Includes memory checks to ensure robust operation

4. LED & Gas Sensor Node Code Overview

The ESP8266 "LED node" controls an LED and monitors a gas sensor:

Setup & Connectivity

- Configures pins for the main LED (D2) and gas indicator LEDs (D1 green, D3 red)
- Establishes Wi-Fi connection

LED Control

- Polls /in-name/led/la every 2 seconds
- Parses "ON"/"OFF" commands and toggles the LED only on change

Gas Monitoring

- Reads analog gas value
- If value > 400, turns on red LED and posts the value to /in-name/gas_sensor/data

5. Fan & Solenoid Node Code Overview

This ESP8266 node mirrors the LED node but handles fan speed and solenoid lock:

Setup & Connectivity

- Assigns GPIOs for solenoid lock and PWM fan output
- Connects to Wi-Fi

Command Fetching

• Polls /in-name/fan/la and /in-name/solenoid/la endpoints every 2 seconds

Actuation Logic

- Fan: Parses a 0-3 speed value and sets PWM duty cycle (0 = off, 1-3 = low/med/high)
- Solenoid: Controls door lock based on "LOCK"/"UNLOCK" commands

State Tracking

- · Remembers last commands to avoid redundant writes
- Logs actions for debugging

6. Workflow Description

- 1 Wearable Fall & Voice Event: ESP32 detects a fall or touch, records audio or fall data, and uploads to OM2M
- 2 Backend Processing: Python script polls OM2M, reassembles audio, transcribes via FasterWhisper, semantically matches commands
- 3 Command Dispatch: Backend sends a content instance back to the relevant OM2M container (LED, fan, solenoid)
- 4 **Node Actuation:** ESP8266 nodes poll their respective containers, parse commands, and actuate hardware accordingly
- 5 **Data Persistence**: A dedicated Python script fetches OM2M entries and stores them in a local database

7. Challenges Encountered

- Microphone Calibration & Noise: Original sound sensor failed to capture clear voice; replaced with MAX9814 for better quality
- Fall Sensor Calibration: Tuning free-fall and impact thresholds on the MPU6050 was challenging;
 required extensive testing
- Back-Voltage on Actuator Nodes: Fan and solenoid back EMF caused spurious triggers; addressed with diodes and filtering capacitors
- Audio Chunking & Memory Constraints: ESP32 RAM couldn't hold full audio; implemented multi-part

transmission solution

• Semantic Command Matching: Adjusting the SentenceTransformer similarity threshold to 0.2 balanced flexibility with accuracy

8. Future Improvements

- Add a web/mobile dashboard for live monitoring and manual override
- Enable dynamic addition of new voice commands at runtime
- Integrate noise-reduction preprocessing on the backend for cleaner transcriptions
- Provide audio/LED feedback upon successful command execution at each node

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