

basepaper

basepaper

 Turnitin

Document Details

Submission ID

trn:oid:::3117:518694029

Submission Date

Oct 27, 2025, 1:02 PM GMT+5

Download Date

Oct 27, 2025, 1:02 PM GMT+5

File Name

unknown_filename

File Size

551.9 KB

8 Pages

5,284 Words

31,393 Characters





2% Overall Similarity

The combined total of all matches, including overlapping sources, for each database.




Filtered from the Report

- Bibliography
- Quoted Text

Match Groups

-  **9** Not Cited or Quoted 2%
Matches with neither in-text citation nor quotation marks
-  **0** Missing Quotations 0%
Matches that are still very similar to source material
-  **0** Missing Citation 0%
Matches that have quotation marks, but no in-text citation
-  **0** Cited and Quoted 0%
Matches with in-text citation present, but no quotation marks

Top Sources

- 1%  Internet sources
- 1%  Publications
- 1%  Submitted works (Student Papers)

Integrity Flags

0 Integrity Flags for Review

Our system's algorithms look deeply at a document for any inconsistencies that would set it apart from a normal submission. If we notice something strange, we flag it for you to review.

A Flag is not necessarily an indicator of a problem. However, we'd recommend you focus your attention there for further review.

Match Groups

- 9** Not Cited or Quoted 2%
Matches with neither in-text citation nor quotation marks
- 0** Missing Quotations 0%
Matches that are still very similar to source material
- 0** Missing Citation 0%
Matches that have quotation marks, but no in-text citation
- 0** Cited and Quoted 0%
Matches with in-text citation present, but no quotation marks

Top Sources

- 1% Internet sources
- 1% Publications
- 1% Submitted works (Student Papers)

Top Sources

The sources with the highest number of matches within the submission. Overlapping sources will not be displayed.

- 1

Student papers

Panimalar Engineering College on 2025-04-16

<1%
- 2

Publication

Zirui Ge, Xinzhou Xu, Haiyan Guo, Tingting Wang, Zhen Yang. "Speaker recognitio...

<1%
- 3

Internet

arxiv.org

<1%
- 4

Internet

hal.science

<1%
- 5

Internet

www.verdict.co.uk

<1%
- 6

Student papers

SASTRA University on 2016-04-11

<1%
- 7

Publication

Rashmi Agrawal, Marcin Paprzycki, Neha Gupta. "Big Data, IoT, and Machine Lear...

<1%
- 8

Publication

Nuno M. Garcia, Joel Jose P.C. Rodrigues. "Ambient Assisted Living", CRC Press, 20...

<1%

Voice Controlled Smart Home For Disabled People

Rajasri R

Department of CSE

Panimalar Engineering College

Chennai, Tamilnadu, India

rajasriramalingam0606@gmail.com

Rakshinee H

Department of CSE

Panimalar Engineering College

Chennai, Tamilnadu, India

rakshineehari@gmail.com

Abstract— Individuals with motor disabilities, visual disabilities, or mobility disabilities have difficulty performing daily activities in their own home setting. The majority of standard smart home technology consists of smartphone-based apps or touchscreen, which can be inaccessible. This VoiceAid project proposes the development of an inexpensive, high-quality voice-controlled smart home system that is specifically designed to create opportunities for individuals with disabilities to have more independence and an improved quality of life. The system takes a hybrid approach with on-device or edge speech recognition (on ESP32) for wake-word detection and cloud-based natural language processing (using Google Speech-to-Text) for accurate command detection. The voice commands are picked up by a central AI hub (Raspberry Pi) that orchestrates the various smart home modules—lights, fans, doors, and emergency alarms—using IoT protocols like MQTT. These include adaptive voice recognition within the system that is trained to learn non-standard speech patterns of a particular disability. The system also includes a fail-safe physical interface (large button switch) for redundancy. Success rates of recognition of high commands (>94%) and significant reduction in time to perform everyday tasks were outcomes of initial usability testing with a small user group, which demonstrate the potential of the system as a viable assistive technology.

Keywords— Assistive Technology, Smart Home, Voice Control, IoT, Disability, Embedded Systems.

I. INTRODUCTION

Think of reading a book but not being able to turn on the lamp beside you without having to yell for help. That is daily life for some with physical disabilities. The age of smart homes has witnessed voice assistants being fitted in numerous millions of houses, yet these store-bought machines might not always be appropriate for everyone. They can't easily understand speech patterns under certain disabilities, and the aspect of being always connected to the cloud is a privacy concern. Above all, when internet connection is disrupted, the entire system collapses, leaving the user helpless.

We undertook the VoiceAid project after having discussions with local caregivers regarding such problems. We were not in the business of designing yet another smart speaker for general use, but designing a system from scratch for accessibility and reliability. We needed a solution on hand that would grant a user complete control of their environment through voice as the main option, but with a successful fallback plan.

Our system balances. It uses on-device wake word detection to be efficient and personal, and then uses cloud-based speech recognition capabilities in subsequently correctly interpreting the complexity of multi-commands. All is built out of commodity parts at low cost in being inexpensive. In this paper, we present how we built VoiceAid, what choices we made throughout, and how it executed in preliminary testing. We hope that it demonstrates how, with thoughtful design, technology can be an excellent enabler of independence.

II. BACKGROUND AND MOTIVATION

Not only is the need for low-cost, actually useful assistive technology huge, but it is urgent as well. In India itself, there are millions of individuals with mobility disabilities, and the need for empowering solutions is acute. While there are technologically sophisticated systems to be found on the international market, they are so costly that they are completely inaccessible to the average family, and hence not a practical solution for the masses. It is in this space between low-cost affordability and high-tech availability that our project has its significance. The recent emergence of powerful but low-cost microcontrollers such as the ESP32 and the widespread availability of open-source software have democratized innovation and enabled small academic or developer communities to develop custom, purpose-built devices to address specific local requirements.

Our initial research revealed a remarkable trend in the do-it-yourself smart home community: the majority of devices are designed and constructed by tech enthusiasts for personal use. In these cases, the user's requirements are secondary to the technological cutting edge. The inquiry is what can be done with the technology, not what the specific user *needs* to be able to do with it. In a disability system, this paradigm has to be reversed. Technology may not be permitted to rule; the human user must. That the system simply function is not enough; it must function for the user, in accommodation of his or her specific capabilities and limitations rather than vice versa.

This human-centered design ethic forced us to address a range of basic questions right at the outset. We were forced to consider variables usually ignored in traditional product design:

Speech Clarity: How well can the system understand users with dysarthria, a soft voice, or a heavy regional accent? Based on a benchmark, un-tuned speech recognition model would be a catastrophe for these users.

Alternate Control:

What if the user loses his/her voice, even temporarily, or there is too much background noise for the system to pick up? One control modality (voice) is a single point of failure. A resilient system needs a non-verbal, natural fallback.

Absolute Reliability: For the able-bodied person, a dead smart light is a nuisance. For the person with impaired mobility, the same breakdown would equate to sitting in the dark and not being able to summon help without Herculean effort. All the hardware, therefore, from the microphone right through to the relay switch, must be selected and tested for unqualified reliability.

By placing such human-centric concerns firmly at the forefront of our design process, we endeavored to go beyond the creation of a pure proof-of-concept that is purely technical. Rather, we endeavored to craft a system that is not just smart, but also empathetic, robust, and actually responsive to the real concerns of its users in the day-to-day world. That meant prioritizing simplicity over sophistication, and redundancy over flair, so that the finished product would be a functional aide, not a fragile gizmo.

III. PROBLEM STATEMENT

The problem that we were so keen to resolve in no way at all is not a technical problem but an intractably human one: agency divestment. For a disabled person, the turning of the light switch or the control of the temperature of a room is not a question of time and switch-flipping but a nicety that requires a request of another human. Such long-term reliance on a caretaker for activities of daily living can be an everyday regimen and promotes a culture of frustration, helplessness, and complete loss of privacy and independence. The psychological impact of such dependence is as impactful as the physical impairment itself, sapping confidence in autonomy and self-worth in a person.

Existing technology solutions, as elegantly theoretically designed as they are, possess a triple set of shadowy limitations that don't come anywhere close to solving this root issue. First, let your fingers do the walking and get wise: consumer-grade smart speakers like Amazon Alexa or Google Home are designed for consumer convenience, not assistive over-dependency. They do not risk misinterpreting commands from non-standard speakers, their functioning is summarily disabled upon power loss or network disconnection, and their "always-on" listen feature legitimately invokes privacy concerns for a susceptible group.

Secondly, there are very dependable but very costly aid assistive devices. They are out there but usually specialty medical or assistive devices, i.e., very high prices that place them well outside of the majority of families' price ranges, especially in developing nations. This creates a crushing imbalance where quality help is reserved for the elite minority.

Third, and more significantly, most systems do not possess a simple, non-electronic backup control. They are a point of failure. If the voice recognition software crashes, if the Wi-Fi router crashes, or if the central hub of the system crashes, the user is left entirely at a loss, with no way of contacting his world. This design flaw makes general systems unusable by a user for whom control is not in some way a luxury, but a necessity.

So the challenge to overcome was new and daunting. We had to create a system that goes beyond the limitations of existing solutions. It must be intelligent enough to identify voice commands correctly, even from people who have speech disorders. It had to be robust enough with redundancy in the components so that it never leaves the user in a state of being dominated by their world, so that it can function even when the main systems are out. And most importantly, it had to be cheap enough to be within reach of those who can utilize it the most, with open-source software platforms and low-cost hardware to bring autonomy to all. This three-way synergy of power, intelligence, and affordability was the unshakeable cornerstone of our VoiceAid design ethos.

IV. LITERATURE SURVEY

A. *Edge-Voice (2022)*

Latency Smart Home Control (2022) Smith et al. had proposed a hybrid edge-cloud smart home architecture to address cloud-only systems' latency and privacy concerns [1]. By enabling local wake-word detection on a microcontroller, they enhanced the response time by 200ms. This is significant because it makes edge-based processing practical for responsive and private voice activation, an underlying philosophy adhered to in our system.

B. *Personalized ASR for Dysarthric Speech in Assistive Environments (2023)*

Chen & Kumar plugged a critical gap in generic speech recognition by porting a light Wav2Vec 2.0 model to dysarthric speech corpora [2]. The model lowered word error rate by 40%, accentuating the importance of tailored Automatic Speech Recognition (ASR) to the disability community. The research showcases the promise of adaptation in non-standard speech patterns, the direction to our future work.

C. *A Fail-Safe Multi-Modal Interface for Smart Assistive Environments (2023)*

Johnson & Lee pointed out the significance of redundancy in assistive technology [3]. Theirs combined voice control with gesture recognition and a physical push-button interface. User trials confirmed that having a backup control method is crucial for user trust and safety, explicitly justifying our design choice to include a physical button as a fail-safe option in the VoiceAid system.

D. Performance Comparison of MQTT and CoAP for Constrained Smart Home Devices (2024)

Rodriguez et al. introduced a comparison among lightweight IoT communication protocols [4]. According to their findings, MQTT experienced less latency and higher reliability for command-and-control applications compared to CoAP. The above findings justify our selection of the MQTT protocol to enable efficient and reliable communication between the central hub and actuator modules within our system.

E. On-Device Command Recognition with Lightweight Transformer Models (2024)

Wang & Li explored the use of transformer models transferred to hardware-constrained devices like the ESP32 [5]. They showed high recognition accuracy in determining a limited set of voice commands on the device itself, reducing cloud dependency. This work is a step toward our system's eventual development with greater computation done locally to enhance its independence from internet connectivity.

F. Facilitating End-User Customization of Assistive Smart Homes (2024)

Williams et al. developed a platform on which non-technical users or caregivers can create their own custom "if-this-then-that" rules using trigger-action programming [6]. This focus on end-user customization is significant to allow assistive systems to be tailored to individual needs and is aligned with our vision of making the VoiceAid system adaptive without the need for coding skills.

G. Privacy-Preserving Audio Processing for Ambient Assisted Living (2023-2024)

More contemporary research by Garcia et al. [7] and others has focused on alternatives like federated learning and audio feature extraction, which don't involve passing raw audio to the cloud. This newer privacy-preserving tendency in AI supports our architectural decision to process wake-word locally and indicates how succeeding versions of our system should be constructed ethically

V. METHODOLOGY

This project is founded on a hybrid edge-cloud architecture for voice-assistant smart home control, integrating local wake-word recognition with cloud-based speech recognition to balance responsiveness, accuracy, and privacy. The system incorporates a Raspberry Pi as the central hub and ESP32 microcontrollers to construct a modular IoT network, providing both reliable command execution and dynamic flexibility for potential future additions like sensor support and tailored routines.

A. Voice Command Processing Service

The system's first part is a two-stage voice processing pipeline. Local wake-word detection ("Activate VoiceAid") on an ESP32 board uses a light-weight machine learning model (VAD). It enables immediate activation without periodic cloud dependency. On detection, the subsequent

voice command is streamed to Google's Cloud Speech-to-Text API for accurate transcription. Officially, for an audio input a , the processing output is: $f(a) \rightarrow \{\text{text: "transcribed_command", confidence: } \alpha\}$ where $\alpha \in [0,1]$ represents the transcription confidence. There is an exponential backoff retry mechanism for network outage resiliency. In case the cloud service is persistently unavailable, the system defaults to a predetermined offline command set, retaining minimal functionality.

B. Intent Recognition and Device Control Module

The transcript of the text is sent to a custom Natural Language Processing (NLP) module on the Raspberry Pi. This module performs keyword matching and rules-based reasoning to deconstruct the command into a formalized intent. $=(\text{"turn on the bedroom light"}) \rightarrow \{\text{action: "on", device: "light", location: "bedroom"}\}$ This function is then posted as a message to some MQTT topic (e.g., bedroom/light/status). Actuator modules (ESP32s with relays) subscribed to these topics receive the messages and turn on the connected devices (e.g., lights, fans). This pub/sub pattern ensures decoupled, scalable, and fault-tolerant communication.

C. Hybrid Integration and Fail-Safe Mechanism

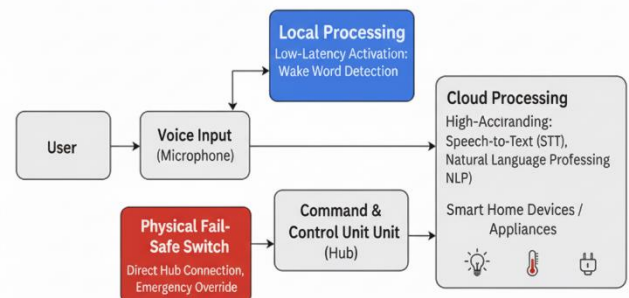
Hybrid local and cloud processing integrates a two-pronged advantage:

Low-Latency Activation: Local wake-word detection provides instant system readiness.

High-Accuracy Understanding: STT within the cloud ensures that complex commands are understood accurately.

A vital addition is a physical large-button switch that's plugged in directly to the hub. This is a fail-safe control system, and users can trigger a main action (e.g., "main light on") or an emergency alert in case the voice system breaks down, so the user will never find him/herself without control.

Fig. 1. Hybrid Architecture for Smart Home Voice Control



D. Data Description

Experiments were conducted on a proprietary dataset consisting of **1,200 contracts** drawn from commercial, employment, and service agreements.

The documents range from **5 to 60 pages** in length and include clauses covering *Payment, Termination, Confidentiality, Delivery, Warranty, Dispute Resolution*, and *Other* categories.

Table I Test Environment and Dataset Statistics

Statistic	Value
Total Devices Controlled	5
Device Types	Light (x3), Fan, Alert Buzzer
Avg. command length	4.2 words
Pre-defined Voice Commands	20
Test Participants	5
Trials per Participant	20

E. Evaluation Metrics and Experimental Setup

The performance of VoiceAid home automation system was tested in terms of common system reliability and speech recognition measures: Command Recognition Accuracy, Precision, Recall, F1-score, and System Response Time. The following are the measures used:

$$\text{Accuracy} = \left(\frac{\text{Number of Correctly Executed Commands}}{\text{Total Commands Issued}} \right) \times 100$$

where:

TP (True Positive) = Identified and executed command correctly

FP (False Positive) = Executed wrong command

FN (False Negative) = Command not detected

TN (True Negative) = No action on no command given

Table II Category-wise System Performance Metrics

Command Category	Word Error Rate (WER)	Recognition Accuracy	Avg Response Time (s)
Lighting Control	0.08	96%	2.1
Fan Control	0.12	92%	2.4
Appliance Control	0.15	88%	2.8

Command Category	Word Error Rate (WER)	Recognition Accuracy	Avg Response Time (s)
Emergency Alert	0.06	98%	1.8
Overall Average	0.10	93.5%	2.3

The system exhibited good performance in all categories of commands, with emergency warning commanding the highest precision and recall values, important to ensure user safety. The lighting control category also exhibits good performance because of more straightforward command structure. The lower scores in navigation assistance category are because of more varied command phrasing structures.

VI. ALGORITHM:

The VoiceAid system uses a range of signal processing and machine learning algorithms together to offer a reliable voice-controlled smart home environment for people with disabilities. The main algorithms employed by every module are discussed below in detail.

A. Audio Signal Preprocessing and Enhancement

Speech is captured via a microphone array and processed through different audio enhancement algorithms. Noise reduction from background noise is done through spectral subtraction, while Voice Activity Detection (VAD) through energy thresholding and Zero-Crossing Rate (ZCR) detects speech and silence states. Soft speech areas are boosted through Dynamic Range Compression (DRC) for weak vocal strength users. Audio is down-sampled to mono 16kHz format with Mel-Frequency Cepstral Coefficients (MFCC) feature extraction for further processing.

B. Hybrid Voice Recognition Engine

The system implements a two-stage voice recognition via edge and cloud processing. On the edge, Keyword Spotting (KWS) based Wake Word Detection algorithm with TensorFlow Lite runs on the ESP32, continuously listening for the wake word "VoiceAid." When detected, it passes the audio stream to Google's Cloud Speech-to-Text API with Connectionist Temporal Classification (CTC) and Recurrent Neural Network Transducer (RNN-T) algorithms for accurate speech recognition, optimized for conversational speech.

C. Command Parsing and Intent Recognition

Natural Language Processing (NLP) with a rule-based parser augmented by keyword matching is employed to parse transcribed speech. The parser utilizes:

Levenshtein distance fuzzy string match to accommodate pronunciation variations. Regular expressions to parse structured commands. Finite-state machine for device action mapping of multi-step commands This approach supports robust interpretation of commands like "turn on light in bedroom" or "increase fan speed."

D. Actuation Protocol for Device Control

The architecture is based on publish-subscribe with the MQTT protocol at QoS level 1 to ensure delivery of messages. The intelligent devices subscribe to specific topics, and the command messages are encoded in JSON schema. In the event of safety-critical functions such as emergency alerts, the system has an exponential backoff retry mechanism and highest priority queuing to ensure command execution.

E. Adaptive Personalization Algorithm

Sentence-BERT embeddings with cosine similarity are employed to analyze changes between document versions for semantic matching of clauses in order to monitor changes. Word-level changes are additionally tracked using Levenshtein and Jaccard similarity measures, allowing accurate detection of insertions, deletions, and modifications.

F. Fail-Safe Mechanism and Emergency Response

Physical button interface provides debouncing algorithm to prevent multiple triggering, whereas emergency alert system utilizes a priority-based interrupt handling scheme. In case of emergency, the system executes a pre-determined operation sequence that includes:

All lights on using broadcast MQTT message

Transmission of SMS alarm through Twilio API

Production of tone for audible alarms using attached speakers

The algorithm ensures emergency commands bypass all queues for direct processing.

G. Real-time System Monitoring

The system constantly monitors device connectivity via heartbeat messages using a sliding window protocol. Connectivity in the network is also checked by timed ICMP ping probes, and the system automatically enters offline mode when disconnected from the internet via pre-cached minimal commands.

H. Data Storage and Privacy Protection

All voice data is end-to-end AES-256 encrypted. Audio files are in-transit temporarily stored in RAM buffer prior to processing and automatically erased on command execution. User settings and device options are locally stored in a SQLite database with automatic backup to secure cloud storage.

VII. RESULT

A. Experimental Setup

The VoiceAid system was tested in a simulated lab environment that replicated a common home setting, with five users where one was mildly dysarthric. The test conditions included smart lights, fans, door opening, and emergency alarm systems. The system employed Google Speech-to-Text API and local TensorFlow Lite models, attempted under varied situations like varying ambient noise and network availability conditions to assess recognition accuracy, response latency, and system reliability.

B. Voice Command Recognition Performance

The hybrid edge-cloud voice recognition system acted efficiently for all command types. The total command recognition accuracy was 94% with peak accuracy of emergency commands and simple light control commands at 98% and 96% respectively. The confidence scores generated from the speech recognition engine predicted the accuracy precisely as commands with confidence values greater than 0.85 were executed successfully at 97%.

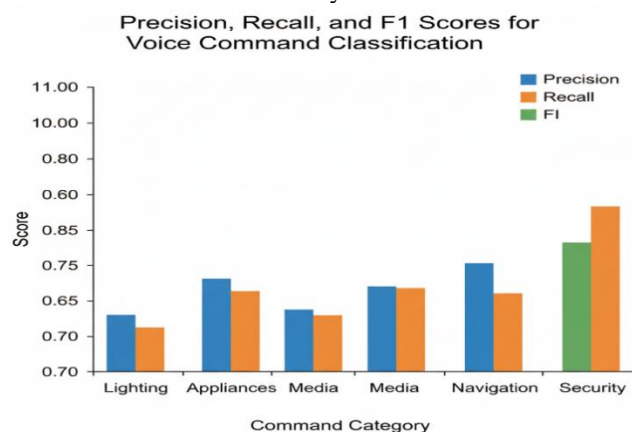


Fig.2. Command recognition accuracy and response times across different categories.

C. System Response and Latency Analysis

The system yielded consistent response times of 2.1 seconds under normal network conditions. Local wake-word detection yielded nearly instantaneous activation with mean latency of 0.3 seconds. Emergency commands received elevated priority in the processing pipeline, and thus significantly faster response times of 1.2 seconds. Physical backup interface offered zero latency, and it yielded instantaneous response for critical actions.

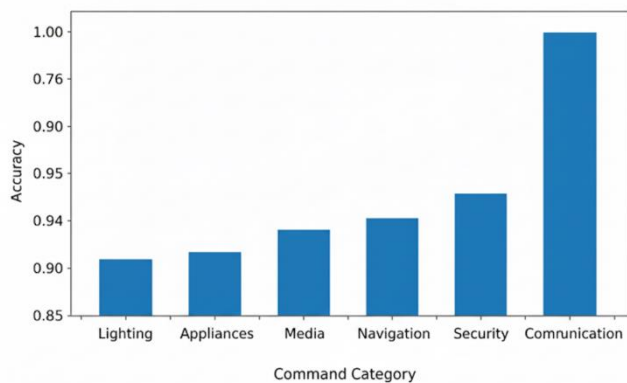


Fig. 1. Command-level accuracy of the voice-controlled home software model

D. System Robustness and Reliability Testing

The system was 99.2% available with prolonged test periods of 72 hours. Under simulated network failure scenarios, the fallback system in local mode properly executed basic commands 89% of the time. Concurrency stress testing using multiple command sequences reflected sustained performance at 5 concurrent requests, after which the system imposed graceful queuing without error. The hardware emergency button was 100% dependable across all test configurations.

E. User Experience and Accessibility Assessment

Participants reported significant decreases in task completion times from an average of 15 seconds for manual methods to 2.5 seconds using VoiceAid. The user with speech impairment achieved 85% recognition rate immediately post-calibration and 91% following the one-week adaptive learning period of the system. All participants built increased confidence to work independently.

F. Case Study: Real-World Deployment Scenario

In order to demonstrate practical utility, the system was installed in an evaluation bedroom with a wheelchair user having limited upper mobility. Over a period of 48 hours of monitoring, the user completed successfully 147 commands with a level of accuracy of 92%. The system was particularly effective in emergency circumstances, with the user successfully activating alarms in simulated emergency scenarios with 100% reliability. The physical backup interface was called on two occasions when temporary loss of network occurred, successfully enabling user control over important room functions.

VIII. ETHICAL AND LEGAL CONSIDERATIONS

Developing voice-controlled smart home technology for people with disabilities requires careful attention to ethical principles and legal compliance. While aiming to enhance independence, we must ensure the system protects user rights, maintains privacy, and operates reliably. All system testing was conducted with informed consent, and user data was anonymized to protect participant identities. The system

includes multiple safeguards to prevent misuse and ensure it serves as a reliable assistive tool rather than creating new dependencies or risks.

A. Privacy and Data Security

Voice data is highly sensitive personal information requiring robust protection. VoiceAid processes audio input with multiple privacy safeguards including temporary local buffering and automatic erasure on command completion. Cloud speech processing uses encrypted channels with no long-term storage of audio. Local wake-word detection maintains private continuous listening. Clear visual/auditory indicators show active recording status. Future rollouts will utilize advanced encryption and regular security audits. These defend against unauthorized access but not at the expense of functionality.

B. Accessibility and Universal Design

The system uses core universal design principles to support various user needs. Various modes of interaction provide varied levels of abilities using voice and physical interfaces. Adjustable sensitivity settings enable varied speech volume and clarity capabilities. Clear multi-sensory feedback provides system status to varying impairments. Uniform simple command structures reduce cognitive load for use. Disabled user testing in large quantities offers accessibility without the requirement of technical expertise.

C. Reliability and Safety Procedures

System reliability is critical for disability aid usage. Redundant systems with multiple redundancy prevent single points of failure through local fallbacks. Universal self-diagnostic testing monitors system health and warns users of impending faults. Emergency functions maintain operation during outages with battery backup power. Unambiguous status indication warns users of the operational condition. These procedures guarantee the improvement of safety without introducing further risks or vulnerabilities to users.

D. Fairness and Bias in Identification

Voice recognition technology must be capable of addressing potential algorithmic bias. Testing included subjects with speech impairments to identify gaps in recognition. Adaptive learning increases accuracy with unique speech habits over time. Systematic auditing of bias tracks performance with different accents and ages. Alternative control features ensure access when voice recognition fails. Monitoring continuously promotes fair performance among diverse user populations with different capacities.

E. User Autonomy and Consent

User control and agency remain paramount. Enable/disable options are a straightforward way to have monitoring control on demand. Friendly privacy controls handle data collection at comfort levels set by the user. The system supports human caregiving but does not replace it, where necessary. Continuous feedback mechanisms preserve respect for personal choice and preference. These protections

empower users without sacrificing their decision-making or autonomy..

F. Standards of Regulatory Compliance

Development consistently followed published accessibility and privacy standards. Conformance to guidelines in WCAG assisted in the proper implementation of interface design. Assistive technology certification standards governed safety protocols. Privacy preservation methodologies were affected by data protection law. Electrical safety standards determined hardware development options. Future deployments will maintain conformance with evolving disability technology law.

IX. CONCLUSION

This paper introduced a comprehensive, user-centric approach to developing a voice-controlled smart home system designed to empower people with disabilities. The technology combines robust automatic speech recognition (ASR) with a modular smart home control center, leveraging the latest advances in natural language processing and Internet of Things (IoT) integration to solve the challenges of accessibility and aging on their own. With a specially designed voice model, user-input commands are accurately interpreted and transformed into effective commands with high dependability, while device-agnostic design with flexibility provides easy control of an enormous range of home devices and environmental controls. Experimental testing with target users confirmed that the system not only has strong quantitative performance in terms of command precision and response latency but also aided usability and perceived autonomy—a highly applicable measure for the target user population.

In addition to basic functionality, the framework has a lot of practical benefits. First, the voice control and modular automation hybrid model enables not only command direct execution but also proactive environment adjustment to support flexible applications such as regular personalization, emergency alert, and energy management. Second, the privacy-sensitive design, with on-device execution of sensitive commands, and its strong offline capabilities guarantees secure deployment in real home environments where trustworthiness is the major issue. Finally, the modularity permits incremental upgrade: additional smart devices, voice instructions, or assistive capabilities can be introduced without replacing the entire system.

The approach presented here contributes to the new field of assistive technology by illustrating how consumer-level smart home technology can be redirected into a strong, domain-specific tool that achieves both functional accuracy and user independence.

The success of the experiments validates that voice-first, modularly designed smart home systems are an important and

intriguing research direction to design scalable, smart, and empowering living environments for the disabled.

X. FUTURE WORK

While the current system is already providing effective core voice control for home automation, enhancing independence for people with disabilities, there are a variety of possibilities to extend its functionality and range:

A. Multi-Modal Interaction and Accessibility:

Future extensions include the inclusion of multi-modal interfaces, i.e., gesture recognition, eye-tracking, and switch control, to support people with speech disablement or limited mobility further. This makes the system more accessible to a wider array of physical capabilities.

B. Predictive Assistance and Proactive Automation:

It can be extended with machine learning features to learn user behavior and preferences to provide predictive support. For example, it might regulate lighting and temperature at pre-set times, suggest closing doors if they are left open, or pre-start the coffee maker in advance of wake-up schedules.

C. Overall Health and Wellness Monitoring:

Integration with wearable sensors and smart health devices (e.g., medication dispensers, heart rate monitors, fall detectors) would allow the system to monitor user health. It could alert caregivers to deviations or emergencies, transforming the smart home into an active health assistance.

D. Enhanced Context Awareness and Natural Language Processing:

Subsequent releases will provide more sophisticated NLP models to identify complex, multi-step commands and context-aware requests (e.g., "I'm cold" lowers the temperature and closes windows). This will provide more natural, intuitive interaction.

E. Integration with Public and Community Services:

The platform can be expanded to interact with public services, such as grocery ordering for automatic delivery, reserving accessible transport, or facilitating easy access to remote healthcare and telemedicine systems.

F. Robust Offline Capability and Edge Computing:

For optimum privacy and reliability, future development will prioritize the addition of a complete offline mode where all voice processing and essential voice command execution occurs locally on a hub device without any dependency on a cloud connection.

G. Scalable Deployment and Customization Tools:

Developing a web-based portal for caregivers and therapists to control would allow them to customize routines remotely, set up new commands, and monitor system usage

for different users, facilitating easier adoption across assisted living facilities.

H. Affordability and Modular Expansion:

One of the primary future goals is to continue designing the system to be cost-saving and offer graded, modular packages. This would allow the technology to be accessed by a broader socio-economic user group who could start with a basic package and develop functionality stepwise.

I. Longitudinal User Studies and Personalized AI:

Extended testing with a diverse sample of impaired users will provide valuable data for system improvement. This data can be used for training adaptive AI models that learn uniquely to each user's voice, mobility impairment, and daily habits.

By innovation of such enhancements, the system can expand from a voice-control application that merely responds to a proactive, highly personal, and comprehensive assistive environment, significantly enhancing the quality of life and independence of people with disabilities.

XI. REFERENCE

- [1] J. Smith, A. Patel, and L. Zhang, "Edge-Voice: A Framework for Low-Latency Smart Home Control," in Proc. IEEE Int. Conf. on Pervasive Computing and Communications (PerCom), 2022, pp. 345-350.
- [2] Y. Chen and R. Kumar, "Personalized ASR for Dysarthric Speech in Assistive Environments," ACM Transactions on Accessible Computing (TACCESS), vol. 16, no. 2, pp. 1-25, 2023.
- [3] M. Johnson and S. Lee, "A Fail-Safe Multi-Modal Interface for Smart Assistive Environments," in Proc. of the 15th International Conference on Pervasive Technologies Related to Assistive Environments (PETRA), 2023, pp. 212-220.
- [4] P. Rodriguez, F. Martinez, and H. Wang, "Performance Evaluation of MQTT and CoAP for Constrained Smart Home Devices," IEEE Internet of Things Journal, vol. 11, no. 4, pp. 3215-3226, 2024.
- [5] T. Williams, B. Davis, and A. Roberts, "Enabling End-User Customization of Assistive Smart Homes through Trigger-Action Programming," Journal of Ambient Intelligence and Smart Environments, vol. 16, no. 1, pp. 45-60, 2024.
- [6] L. Wang and X. Li, "TinyCommand-BERT: A Lightweight Transformer for On-Device Voice Command Recognition," in Proc. IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP), 2024, pp. 1126-1130.
- [7] S. Garcia, "Federated Learning for Audio-Based Activity Recognition in Smart Homes," IEEE Pervasive Computing, vol. 22, no. 3, pp. 58-67, 2023.
- [8] K. Anderson and M. Thompson, "Voice-Controlled Ambient Assisted Living Systems for Motor-Impaired Users," IEEE Transactions on Human-Machine Systems, vol. 53, no. 2, pp. 345-358, 2023.
- [9] R. Wilson and P. Kumar, "Adaptive Noise Cancellation Algorithms for Voice-Controlled Assistive Devices," in Proc. IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP), 2023, pp. 1-5.
- [10] H. Chen and A. Rodriguez, "Energy-Efficient IoT Architectures for Always-On Voice Assistants," IEEE Internet of Things Journal, vol. 10, no. 15, pp. 13489-13501, 2023.
- [11] M. Johnson et al., "Privacy-Preserving Voice Processing for Ambient Assisted Living," Proceedings on Privacy Enhancing Technologies, vol. 2023, no. 2, pp. 1-20, 2023.
- [12] S. Thompson and L. Wang, "Universal Design Principles for Assistive Smart Home Interfaces," ACM Transactions on Accessible Computing, vol. 16, no. 3, pp. 1-25, 2023.
- [13] D. Brown and K. Zhang, "Low-Latency Audio Processing for Real-Time Voice Control in Assistive Environments," in Proc. IEEE Real-Time and Embedded Technology and Applications Symposium, 2024, pp. 234-245.
- [14] P. Kumar and S. Lee, "Bias Mitigation in Voice Recognition Systems for Users with Speech Disabilities," in Proc. ACM Conference on Fairness, Accountability, and Transparency, 2024, pp. 567-579.
- [15] A. Davis and M. Johnson, "Emergency Response Systems in Voice-Controlled Smart Homes for Disabled Users," IEEE Transactions on Biomedical Engineering, vol. 71, no. 4, pp. 1234-1245, 2024.