Voice Biometric Authentication System – Technical Report

1. Introduction

This system implements a voice-based biometric authentication mechanism using Python. It consists of three main scripts:

- register.py: Registers a user's voice.
- authenticate.py: Authenticates a user against previously registered voice data.
- voice_utils.py: Provides reusable utility functions for audio processing, feature extraction, and data management.

The system utilizes Mel-frequency cepstral coefficients (MFCC) for feature extraction and cosine similarity for matching embeddings. SQLite is used for persistent storage of user voice profiles.

2. System Components and Workflow

2.1 Registration Module (register.py)

Purpose: Capture and store a user's voiceprint (audio embedding) for future authentication.

Workflow:

1. Database Initialization:

Calls init_db() to create the necessary SQLite schema if it does not exist.

2. User Input:

- Accepts and validates a username.
- o Terminates if input is empty.

3. Audio Capture:

• Records a 4-second audio sample via record_audio(user).

4. Preprocessing:

- Loads audio data from a .npy file using load_audio_from_npy().
- Performs a liveness check using is_live(audio). If the check fails, the process terminates.
- Applies basic noise suppression with denoise_audio().

5. Feature Extraction:

 Extracts MFCC features via extract_features() and averages across the time axis to obtain a fixed-size embedding.

6. Data Storage:

 Embedding is serialized as a float32 byte array and stored in the users table using save_embedding_to_db().

2.2 Authentication Module (authenticate.py)

Purpose: Verify a user's identity by comparing a live audio sample to the registered voiceprint.

Workflow:

1. Database Initialization:

o Initializes the SQLite database.

2. User Identification:

- Prompts for a username.
- Loads the corresponding embedding using load_embedding_from_db().

3. Audio Capture and Preprocessing:

• Records a 4-second voice sample using record_audio().

 Applies the same preprocessing pipeline: liveness check → denoising → feature extraction.

4. Similarity Computation:

 Uses compare_embeddings() to compute the cosine similarity between the stored and current embeddings.

5. Decision Threshold:

○ The user is authenticated if the similarity score \ge 0.75.

6. Visualization:

 Optionally renders a bar chart using plot_score() to indicate the similarity score.

2.3 Utility Module (voice_utils.py)

Purpose: Provides audio processing, feature extraction, database management, and visualization functions.

Key Components:

• Audio Handling:

- record_audio(filename): Captures audio using the sounddevice library and saves it as a .npy file.
- load_audio_from_npy(filename): Loads a NumPy array from the given filename.

• Feature Extraction:

 extract_features(audio): Extracts 13-dimensional MFCC features and returns their mean vector over time.

• Similarity Measurement:

 compare_embeddings(emb1, emb2): Returns the cosine similarity between two vectors.

• Noise Reduction:

 denoise_audio(audio): Uses a simple spectrogram-based thresholding mechanism to remove low-energy components.

• Liveness Detection:

 is_live(audio): Evaluates signal energy and silence ratio to detect whether the recording appears to be live.

• Database Operations:

- init_db(): Sets up a users table with fields username (TEXT) and embedding (BLOB).
- save_embedding_to_db(username, embedding): Stores the user's voiceprint as a float32 binary blob.
- load_embedding_from_db(username): Retrieves and decodes the stored voiceprint.

• Visualization:

 plot_score(score): Plots a bar chart of the similarity score, using color coding to indicate authentication success or failure.

3. Technical Specifications

Aspect	Specification
Sample Rate	16,000 Hz
Recording Duration	4 seconds
Audio Format	Mono, 32-bit float
Feature Vector Size	13 (MFCC)
Embedding Storage	Serialized NumPy float32 array
Similarity Metric	Cosine similarity
Authentication Threshold	0.75
Storage Backend	SQLite (local)

4. Evaluation

4.1 Strengths

- Modular Architecture: Clearly separated responsibilities across modules.
- Lightweight: Minimal dependencies and fast execution.
- Easy Deployment: No external server or cloud infrastructure required.
- Visual Feedback: Score plotting helps in understanding similarity metrics.

4.2 Limitations

- Liveness Detection:
 - Relatively primitive and could be circumvented by playback attacks.
- Security:
 - Embeddings are stored in plaintext without encryption or access control.
- Noise Handling:
 - Basic denoising approach might not perform well under varied real-world noise conditions.
- Scalability:
 - SQLite is not suitable for large-scale applications or concurrent user access.

5. Recommendations for Enhancement

- Advanced Liveness Detection:
 - Introduce techniques like voice spoofing detection, playback challenge-response, or deep learning-based classifiers.
- Data Security:

- Encrypt stored embeddings.
- Add user authentication for database access.

• Improved Audio Preprocessing:

 Use spectral subtraction, Wiener filters, or neural denoisers for better noise resilience.

Web Integration:

o Develop a RESTful API or web interface for accessibility across devices.

• Evaluation Metrics:

 Add metrics such as False Acceptance Rate (FAR) and False Rejection Rate (FRR) to benchmark system performance.

6. Conclusion

This system provides a functional baseline for voice-based user authentication. Its straightforward design, minimal resource requirements, and modularity make it a good candidate for prototyping or educational purposes. However, deployment in a real-world or production-grade environment would require enhancements in security, scalability, and robustness.