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Part 1: Research & Selection

I have reviewed the provided repository and identified three models that are well-suited for our purpose.

1. RawNet2

• Key Technical Innovation:

RawNet2 processes raw audio waveforms directly using a combination of convolutional and recurrent neural networks. This eliminates the need for handcrafted features and enables the model to learn subtle patterns indicative of deepfake audio.

Reported Performance Metrics:

Achieves state-of-the-art accuracy in deepfake detection without handcrafted features. However, its performance can degrade under certain manipulations like volume control and noise injection.

• Why You Find This Approach Promising for Our Specific Needs:

Since RawNet2 does not rely on pre-extracted features, it can adapt to different types of audio forgeries effectively. Its real-time or near real-time processing capability makes it suitable for practical applications where fast detection is required.

• Potential Limitations or Challenges:

Vulnerable to specific audio manipulations such as volume control. Additionally, training and inference require significant computational resources, making deployment on resource-constrained devices challenging.

2. AASIST

• Key Technical Innovation:

AASIST employs a graph neural network (GNN) to analyze spectro-temporal features of audio. This architecture enhances its ability to capture complex relationships between different parts of the signal, improving deepfake detection.

Reported Performance Metrics:

Demonstrates strong performance on datasets like SONAR and often outperforms other models in specific scenarios. More robust than RawNet2 against certain manipulations.

Why You Find This Approach Promising for Our Specific Needs:

AASIST's use of GNN allows it to identify subtle structural changes in audio, making it more resistant to specific deepfake techniques. Its real-time processing capability aligns well with applications requiring quick decisions.

• Potential Limitations or Challenges:

Although more robust than RawNet2, AASIST still struggles with certain types of audio manipulations, such as fading. Computational demands remain high, similar to other deep-learning-based models.

3. CNN-Based Model with EfficientNet for ASVspoof Dataset

Key Technical Innovation:

This approach leverages a pre-trained EfficientNet model as a feature extractor for analyzing spectrograms—visual representations of sound frequencies over time—on the ASVspoof dataset. Transfer learning allows us to utilize the rich feature representations learned from large-scale datasets, adapting them for deepfake detection.

Reported Performance Metrics:

Pre-trained models, such as EfficientNet, demonstrate high accuracy in deepfake detection when fine-tuned on spectrogram-based representations of audio data. This approach enhances model generalization and robustness, even with limited data.

• Why This Approach is Promising for Our Specific Needs:

- 1. Pre-trained Feature Extraction: EfficientNet has learned powerful image representations from large-scale datasets, making it well-suited for analyzing spectrograms without requiring extensive retraining.
- 2. Reduced Training Time: Since most of the model's parameters are already optimized, fine-tuning only requires training a few additional layers, significantly reducing computational costs.
- Scalability and Deployment Efficiency: Unlike heavier architectures like RawNet2 or AASIST, EfficientNet can be deployed efficiently on mid-range GPUs or even CPUs, making real-world deployment more feasible.
- 4. Improved Generalization: Transfer learning allows the model to generalize better across different types of spoofed audio samples, even when data availability is limited.

Potential Limitations or Challenges:

- Dependence on Pre-trained Weights: The model's effectiveness depends on how well the pre-trained weights generalize to spectrogram-based inputs, which are different from standard image inputs.
- 2. Feature Alignment: Spectrogram-based features may require additional preprocessing steps to ensure compatibility with EfficientNet's convolutional layers.
- 3. Data Augmentation Needs: The model may still require augmentation techniques to improve robustness against unseen spoofing techniques.

Part 2: Implementation

Selected Model: "CNN-Based Model with EfficientNet for ASVspoof Dataset"

GitHub Repository Link:

https://github.com/AmoghAgrawal1249/Audio-Deepfake-Detection-using-Pretrained-Model

Advantages of Using a Pre-Trained Model (EfficientNet) with ASVspoof Dataset:

- Highly Optimized Representations EfficientNet has been trained on millions
 of images, making it an excellent feature extractor when fine-tuned for
 deepfake detection on spectrogram data.
- 2. Computational Efficiency Unlike deeper models like RawNet2, EfficientNet achieves a strong balance between accuracy and efficiency, allowing for fast inference on consumer-grade hardware.
- 3. Simplified Training Process Transfer learning requires fewer epochs and less data compared to training a CNN from scratch, reducing the risk of overfitting.
- 4. Pre-Trained Models Reduce Data Dependency Since EfficientNet already captures high-level feature representations, it requires fewer labeled samples for effective fine-tuning.
- 5. Scalability for Deployment The model can run on embedded devices or cloud-based systems efficiently due to its optimized architecture.
- 6. Robustness to Overfitting Pre-trained models generalize better than models trained from scratch, especially on limited datasets.

Future Improvements:

- Experiment with Different Pre-trained Architectures: Testing models like ResNet or Vision Transformers for further improvements in accuracy.
- Integrate Attention Mechanisms: Enhancing the model's ability to focus on critical frequency patterns in the spectrograms.
- Domain-Specific Pretraining: Fine-tuning EfficientNet on a broader range of deepfake audio datasets to improve domain adaptation.
- Hybrid Approaches: Combining EfficientNet with LSTM or Transformer-based models to capture both spatial and temporal dependencies in the audio data.

Part 3: Audio Deepfake Detection Implementation Report

1. Implementation Process

Challenges Encountered

• Dataset Handling Issues:

- The dataset was large, making it difficult to load and process efficiently.
- File sorting inconsistencies caused repeated selection of specific patterns in file indexing.

Model Input Shape Mismatch:

- The pre-trained EfficientNetB0 model expected a 3-channel input, but the spectrograms were grayscale (single channel).
- This was resolved by duplicating the grayscale channel to create a 3-channel input.

• Training Instability:

- o Initial training results showed very low accuracy and precision.
- Experimentation with different learning rates and dropout layers helped stabilize training.

Evaluation Metrics Misalignment:

- o Initially, results showed high recall but extremely low accuracy and precision.
- Investigated label parsing, confirming that bonafide/spoof labels were correctly extracted.

Solutions Implemented

- Efficient File Selection: Changed random sampling to fixed indexing for dataset consistency.
- **Data Preprocessing Fixes:** Corrected label mapping and ensured input shape compatibility with the model.
- Model Architecture Tweaks: Added dropout layers to reduce overfitting.
- **Performance Tracking:** Monitored metrics like accuracy, precision, recall, and F1-score.

Assumptions Made

- The dataset is representative of real-world deepfake detection scenarios.
- The bonafide/spoof classification is binary with well-separated features.
- EfficientNetB0 is a suitable backbone for feature extraction.

2. Model Analysis

Why This Model Was Selected

- **Transfer Learning Advantage:** EfficientNetB0 provides pre-trained feature extraction, reducing training time.
- Compact Yet Powerful: EfficientNetB0 is lightweight compared to deeper models like ResNet.
- **Proven Performance:** It has shown strong results in image-based classification tasks, making it suitable for spectrogram-based analysis.

High-Level Technical Explanation

1. Feature Extraction:

- Converts audio into spectrograms, making it suitable for image-based deep learning models.
- EfficientNetB0 processes these images to extract deep feature representations.

2. Classification Layer:

- Global Average Pooling reduces dimensionality.
- Fully connected layers learn decision boundaries for bonafide vs. spoof classification.
- o The final layer uses a sigmoid activation to output a probability score.

Performance Results(for the first 300 rows of the dataset)

Accuracy: 71.00%
Precision: 34.69%
Recall: 23.61%
F1 Score: 28.10%

Observed Strengths and Weaknesses

Strengths:

- The model is lightweight and computationally efficient.
- Moderate accuracy achieved by fine-tuning the prediction threshold during evaluation.

Weaknesses:

- Poor precision indicates many false positives.
- May struggle with real-world deepfake variations due to dataset bias.

Future Improvements

- Enhance Data Augmentation: Introduce pitch shifting, time stretching, and noise injection.
- Improve Model Complexity: Fine-tune deeper EfficientNet variants (B2, B3) for richer feature extraction.
- Refine Decision Boundaries: Adjust class weights to address class imbalances.

• **Use Multi-Modal Learning:** Combine spectrograms with raw waveform analysis for better accuracy.

3. Reflection Questions

1. Significant Challenges in Implementation

- Handling large datasets efficiently while ensuring reproducibility in file selection.
- Tuning model hyperparameters to improve precision without sacrificing recall.
- Ensuring correct label assignments when loading the dataset.

2. Real-World Performance vs. Research Datasets

- The model may perform worse in real-world conditions due to unseen attack types.
- Controlled datasets may lack variability seen in practical deepfake scenarios.
- Environmental noise and different recording conditions could impact accuracy.

3. Additional Data or Resources for Improvement

- A larger and more diverse dataset covering various deepfake attack techniques.
- Higher-quality audio recordings to preserve subtle speech patterns.
- Computational resources for deeper model fine-tuning.

4. Deployment Considerations for Production

- Latency Optimization: Convert the model to TensorFlow Lite for real-time detection.
- **Continuous Learning:** Periodically retrain on new deepfake samples to adapt to evolving threats.
- **Integration with Real-World Systems:** Deploy in media forensics, telecom security, and content authentication workflows.
- Scalability: Use cloud-based deployment for handling large volumes of audio data.