# Physical Sensors and Systems for Environmental Signals A Comparative Study of Denoising Techniques for Speech Audio Signals

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# Introduction

### Introduction

- Environmental acoustic recordings are crucial for applications in ecology, urban planning, and environmental monitoring.
- However, these recordings are often contaminated with noise from various sources.
- In this study, we compare classical denoising techniques with modern neural network approaches.
- Datasets: Clean speech from LibriSpeech and noise from UrbanSound8K.

# Related Work

### **Related Work**

#### Classical Methods:

- Spectral Subtraction [1]
- Wiener Filtering [2]

#### Neural Network Methods:

- Residual Autoencoder
- U-Net (UNetSpec)
- Hybrid Denoiser
- Transformer Autoencoder

# Methods

## **Spectral Subtraction**

#### Process:

- 1. **STFT:** Compute the Short-Time Fourier Transform (STFT) of the noisy signal to obtain magnitude and phase.
- 2. **Noise Estimation:** Estimate the noise spectrum (often using initial frames assumed to be noise-dominant).
- Subtraction: Subtract the estimated noise magnitude from the noisy magnitude. Use a max operation to avoid negative values.
- iSTFT: Reconstruct the time-domain signal by applying the inverse STFT (iSTFT) using the original phase.
- **Pros:** Simple and computationally efficient.
- Cons: May introduce "musical noise" artifacts due to imperfect noise estimation.

## Wiener Filtering

• **Principle:** Minimizes the mean squared error (MSE) between the estimated clean signal and the true clean signal.

#### Process:

- 1. Estimate the power spectral density (PSD) of both the clean signal and the noise.
- Calculate the Wiener filter, which balances noise reduction and signal preservation.
- 3. Apply the filter in the time domain to the noisy signal.
- Pros: Statistically optimal under assumptions of stationarity.
- Cons: Performance decreases when noise is non-stationary.

### **Overview of Neural Network Denoisers**

- Residual Autoencoder: Processes raw waveforms in the time domain using residual learning.
- U-Net (UNetSpec): Enhances the magnitude spectrogram (frequency domain) with skip connections.
- Hybrid Denoiser: Combines both time-domain and frequency-domain processing for improved denoising.
- Transformer Autoencoder: Uses a simplified attention mechanism to weigh spectrogram features.

# Residual Autoencoder (Details)

#### Architecture:

- **Encoder:** Series of 1D convolutional layers that extract temporal features.
- **Decoder:** Transposed convolutions to reconstruct the signal.
- Residual Connection: The network predicts the noise component; subtracting it from the input yields the denoised signal.
- Advantage: Direct processing of the raw waveform without domain conversion.

# U-Net (UNetSpec) (Details)

#### Architecture:

- Operates on the magnitude spectrogram obtained from the STFT.
- Uses an encoder-decoder structure with skip connections to preserve fine details.
- Reconstructed magnitude is combined with the original phase for the final signal.
- Advantage: Effective at preserving and enhancing spectral details.

# Hybrid Denoiser (Details)

- Dual-Branch Architecture:
  - Time-Domain Branch: Similar to the Residual Autoencoder.
  - Frequency-Domain Branch: Processes the magnitude spectrogram using a U-Net-like structure.
- **Fusion:** The outputs of both branches are concatenated and merged to produce the final denoised waveform.
- Advantage: Leverages complementary information from both the time and frequency domains.

# **Transformer Autoencoder (Details)**

#### Architecture:

- Converts the time-domain signal to a spectrogram via STFT.
- Incorporates a simplified attention block (channel-wise attention) in the bottleneck.
- The decoder reconstructs the enhanced spectrogram, which is then used with the original phase to recover the waveform.
- Advantage: Provides global feature weighting with lower computational overhead compared to full transformer models.

### **Loss Functions Overview**

- Two training loss variants are used:
  - Simple Loss (v1): A combination of L1 loss and Mean Squared Error (MSE) computed in the time domain.
  - 2. **Hybrid Loss (v2):** Combines time-domain loss, frequency-domain loss, and a negative SI-SDR term.

# Simple Loss (v1)

$$L_{\text{simple}} = \frac{1}{N} \sum_{i=1}^{N} |x_i - \hat{x}_i| + \frac{1}{N} \sum_{i=1}^{N} (x_i - \hat{x}_i)^2$$

- L1 Loss: Penalizes absolute differences; robust to outliers.
- MSE Loss: Emphasizes larger errors.
- The sum of both encourages both overall fidelity and the preservation of fine details.

# Hybrid Loss (v2)

- Components:
  - Time-domain L1 Loss:

$$L_{time} = \frac{1}{N} \sum_{i=1}^{N} |x_i - \hat{x}_i|$$

Frequency-domain L1 Loss:

$$L_{freq} = \frac{1}{FT} \sum_{f=1}^{F} \sum_{t=1}^{T} \left| |X(f,t)| - |\hat{X}(f,t)| \right|$$

- Negative SI-SDR: Optimizes the Scale-Invariant Signal-to-Distortion Ratio.
- Overall Hybrid Loss:

$$L_{\text{hybrid}} = \frac{1}{3} \left( L_{time} + L_{freq} + (-\text{SI-SDR}) \right)$$

 This loss encourages accurate reconstruction in both the time and frequency domains while directly minimizing signal distortion.

# **Metrics**

### **Evaluation Metrics**

### PESQ (Perceptual Evaluation of Speech Quality):

- Measures the perceived quality of speech.
- Scale: Approximately -0.5 to 4.5 (higher scores indicate better quality).

### STOI (Short-Time Objective Intelligibility):

- Assesses the intelligibility of speech.
- Scale: 0 to 1 (values closer to 1 indicate higher intelligibility).

### • SI-SDR (Scale-Invariant Signal-to-Distortion Ratio):

- Evaluates the overall distortion introduced by the denoising process.
- Higher values denote less distortion.

### MOS (Mean Opinion Score):

A subjective measure of audio quality, typically rated from 1 to
5.

# Experimental Setup

## **Experimental Setup**

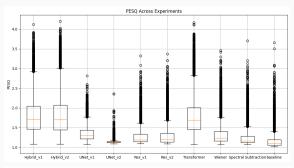
- **Optimizer:** AdamW with a learning rate of 3e-4 and weight decay of 1e-5.
- **Scheduler:** ReduceLROnPlateau to adjust learning rate when the validation loss plateaus.
- **Training:** Batch size of 24 over 10 epochs (demonstration setting; longer training is recommended).
- Dataset: Synthetic noisy data generated by mixing LibriSpeech with UrbanSound8K at various SNR levels.

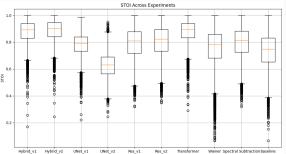
# **Results**

# **Quantitative Results**

Method	PESQ	STOI	SI-SDR (dB)	MOS
Baseline (No Denoising)	$1.16 \pm 0.19$	$0.74 \pm 0.12$	0.10 ± 4.22	2.84 ± 0.82
Spectral Subtraction	$1.25 \pm 0.21$	$0.80\pm0.11$	$3.76 \pm 5.72$	$3.83 \pm 0.54$
Wiener Filtering	$1.33 \pm 0.26$	$0.76 \pm 0.14$	$-0.10 \pm 6.51$	$3.82 \pm 0.47$
ResAutoencoder (v2)	$1.29 \pm 0.24$	$0.81\pm0.11$	$3.18 \pm 5.69$	$2.52 \pm 0.63$
U-Net (v1)	$1.35 \pm 0.18$	$0.78 \pm 0.08$	$3.95 \pm 3.63$	$3.54 \pm 0.78$
Hybrid (v2)	$1.81 \pm 0.50$	$0.89 \pm 0.08$	$11.69 \pm 5.27$	$3.79 \pm 0.74$
Transformer	1.78 ± 0.45	$0.88 \pm 0.08$	11.65 ± 4.90	$2.87 \pm 0.50$

## **Boxplots of Evaluation Metrics**





## **Comparative Conclusions**

- Classical vs. Neural Methods: Neural approaches (especially Hybrid and Transformer models) outperform classical methods in reducing distortion (SI-SDR) and improving intelligibility (STOI).
- Architecture Insights:
  - Hybrid Denoiser: Achieves the highest SI-SDR, indicating minimal distortion.
  - **U-Net:** Excels in preserving spectral details (PESQ and MOS).
  - Transformer: Provides competitive SI-SDR with slightly lower perceptual quality (MOS), suggesting room for further tuning.
- Consistency: Boxplots show that deep learning models not only improve mean performance but also reduce variability.
- **Trade-Offs:** High SI-SDR values must be balanced with perceptual quality, as indicated by MOS.

# **Future Work**

### **Future Work**

- Validate the models on real-world environmental recordings.
- Explore advanced architectures (e.g., full transformer models and GAN-based approaches).
- Develop adaptive and semi-supervised denoising methods.
- Investigate additional evaluation metrics that better capture perceptual quality.

# Conclusion

### **Conclusion**

- Deep learning approaches (Hybrid and Transformer) substantially outperform classical methods.
- Combining time- and frequency-domain information is key for effective denoising.
- Neural models achieve improved signal fidelity and intelligibility, though further tuning is needed for optimal perceptual quality.

Thank you for your attention!

### References i



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