

Modeling ANN in Real Life: Group selects an application, map ANN structure, neuron types, and suitable learning law.

Abstract

Noise interference is a common problem in modern communication and audio systems. Devices such as smartphones, hearing aids, headsets, medical monitoring instruments, and broadcasting systems frequently operate in environments filled with unwanted background sounds. These disturbances degrade signal quality and reduce clarity. Conventional filtering techniques work effectively for predictable and steady noise but often fail when the noise is irregular, time-varying, or nonlinear.

Artificial Neural Networks (ANNs) provide a data-driven solution to this challenge. Instead of relying on predefined mathematical filters, neural networks learn the relationship between noisy inputs and clean signals through training. Their adaptive nature enables them to model complex patterns and dynamically changing noise conditions.

This project presents a conceptual design of an ANN-driven noise reduction model. It explains the selected network architecture, describes the types of neurons involved, and discusses the learning rule used for training. The overall functioning of the system is also detailed to demonstrate how the network progressively improves signal clarity by minimizing prediction error. The study highlights the practical importance of ANN techniques in advanced signal processing applications.

1. Introduction

Signals transmitted in real-life conditions are rarely free from disturbances. During phone conversations, live streaming, music recording, or biomedical signal monitoring, unwanted environmental sounds mix with the useful information. These interferences can originate from vehicles, machinery, wind, electrical devices, or crowd noise.

Noise reduction aims to separate useful signal components from irrelevant disturbances while maintaining the integrity of the original information. Traditional approaches such as low-pass, high-pass, and adaptive filters rely on fixed mathematical models. Although these methods are computationally efficient, they often struggle when noise patterns change unpredictably.

Artificial Neural Networks offer a flexible alternative because they learn directly from data. By training the network using examples of noisy and clean signals, the system learns how to approximate the clean output. This adaptability makes ANN particularly suitable for dynamic environments.

This project focuses on:

- Designing an ANN structure for signal denoising
 - Identifying neuron types used in each layer
 - Selecting an appropriate training rule
 - Explaining how the model operates step by step
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2. Project Goals

The main goals of this study are:

- To analyze how noise affects signal transmission
 - To design a neural network model for reducing noise
 - To determine appropriate activation functions and neuron types
 - To apply a suitable supervised learning mechanism
 - To describe the operational workflow of the proposed system
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3. Fundamentals of Noise Reduction

Noise reduction involves detecting unwanted signal components and minimizing their influence without distorting useful information.

3.1 Passive Approach

Passive methods use insulation materials or structural design to physically block noise. Examples include padded headphones or soundproof walls. These methods are simple but limited in adaptability.

3.2 Active Approach

Active systems electronically generate an anti-noise waveform that cancels unwanted sound via phase inversion. This approach is effective for continuous low-frequency disturbances.

3.3 Neural Network-Based Approach

ANN-based noise reduction enhances active systems by replacing fixed filters with learned models. The network studies patterns in both noisy and clean signals and learns how to reconstruct a purified version of the input. This makes the system flexible and capable of handling non-stationary noise.

4. Neural Network Architecture

The proposed model uses a **multilayer feedforward neural network**, where data flows sequentially from input to output.

4.1 Input Layer

The input layer accepts:

- Noisy signal samples (time-domain values)
- Optional reference noise input

Each neuron corresponds to a numerical feature extracted from the signal. The layer simply forwards these values to deeper layers without modification.

4.2 Hidden Layers

Hidden layers perform the core computations of the system.

Their responsibilities include:

- Learning complex patterns in signal data
- Identifying repetitive or random noise components
- Transforming input features into meaningful internal representations

Nonlinear activation functions are applied to enable the network to model complex relationships.

4.3 Output Layer

The final layer produces the estimated clean signal.

Because noise cancellation is a regression task (continuous output), linear activation is generally used. The output is compared with the actual clean signal during training to compute error.

5. Neuron Categories

5.1 Linear Output Neurons

Used in the final layer to produce smooth continuous values that approximate the clean signal.

5.2 Nonlinear Hidden Neurons

Activation functions such as:

- ReLU (Rectified Linear Unit)

- Sigmoid
- Tanh

These introduce nonlinearity, enabling detection of complex signal patterns.

5.3 Trainable (Adaptive) Neurons

Weights and biases are continuously updated during training to improve prediction accuracy. This adaptability makes the system responsive to varying environments.

6. Learning Mechanism

Supervised Error-Based Learning

The model is trained using pairs of:

- Noisy input signal
- Corresponding clean target signal

The objective is to minimize the prediction error.

Error Calculation

$$Error = Target - Output$$

Weight Update Rule

Weights are adjusted using gradient descent and backpropagation:

$$w_{new} = w_{old} - \eta \frac{\partial E}{\partial w}$$

Where:

- η = Learning rate
 - E = Loss value
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Loss Function

The most common loss function used is Mean Squared Error (MSE):

$$MSE = \frac{1}{n} \sum (y_{true} - y_{pred})^2$$

Lower MSE indicates improved signal reconstruction.

7. Operational Workflow

1. The noisy audio sample enters the network.
2. Hidden layers analyze features using weighted computations.
3. The network predicts a cleaner signal.
4. The predicted signal is compared with the original clean signal.
5. Error is computed.
6. Weights are updated via backpropagation.
7. The cycle repeats until acceptable accuracy is achieved.

Over time, the model becomes capable of suppressing noise effectively.

8. Benefits of ANN-Based Noise Reduction

- Handles nonlinear and unpredictable noise
 - Learns directly from real data
 - Adapts to changing acoustic conditions
 - Suitable for real-time applications with optimized hardware
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9. Challenges

- Requires substantial training data
 - High computational complexity
 - Sensitive to training data quality
 - May require hardware acceleration for real-time use
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10. Practical Applications

- Intelligent headphones
- Smart voice assistants
- Video conferencing tools
- Hearing assistance devices
- Broadcasting and recording studios
- Medical signal enhancement systems

11. Conclusion

This project introduced a conceptual Artificial Neural Network model for noise suppression. By designing a feedforward architecture, selecting appropriate neuron types, and applying supervised error-correction learning, the system effectively reconstructs clean signals from noisy inputs.

The adaptability and learning capacity of ANN make it highly suitable for real-world audio processing applications. This model provides a foundational understanding for future advancements in intelligent signal processing technologies.

12. Future Enhancements

- Implementation using deep learning libraries
 - Use of Convolutional Neural Networks with spectrogram inputs
 - Integration into embedded low-power devices
 - Hybrid models combining ANN with reinforcement learning
 - Exploration of recurrent networks for sequential signal processing
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13. References

- Simon Haykin – Neural Networks: A Comprehensive Foundation
- S. Haykin – Adaptive Signal Processing
- Goodfellow, Bengio & Courville – Deep Learning
- Digital Signal Processing Lecture Materials