

# Classification of Phonetic Syllables Using Stacked Autoencoder and Characterization via Centroid

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**Abstract**—This work presents a new method for layer insertion in stacked autoencoder neural networks. In this approach, a branch of layers is inserted on the side of the last hidden layer, and the output layer after the training of an existing layer has stabilized. Later, the new branch is merged with the previous layers. This insertion type is called collaborative, as it introduces new knowledge to the network without reducing the knowledge already acquired by the previous layers. This approach enables a neural network to learn with reduced design time. It overcomes the typical problem of defining the number of layers and the number of neurons in each layer. This technique was applied in phonetic syllable classification, where the Fast Fourier Transform obtains the audio data. These audio data were processed using a vertical bar plot to compress the audio data by using centroids. This procedure provided data compression without losing characteristics. Thus, collaborative insertions were evaluated in terms of the degree of growth for a multi-class classification problem.

**Index Terms**—Artificial Neural Networks, Deep Learning, Stacked Autoencoder Networks, and Centroids.

## I. INTRODUCTION

Pattern recognition is an interdisciplinary area that combines machine learning and artificial intelligence to detect and classify patterns in complex data [1], [2]. Speech recognition is a specific branch focused on extracting, processing, and categorizing information from speech signals [3], [4]. Despite its global relevance, Portuguese still lacks extensive Automatic speech recognition (ASR) research and resources [5], motivating efforts toward specialized methods for this language.

ASR systems operate through preprocessing, feature extraction, and classification [6]. Preprocessing reduces noise and speaker variability, while feature extraction generates compact representations that strongly affect classification performance [7]. Common representations include LPC, PLP, GFCC, DWT, FBANKs, and MFCC [8], [9].

Artificial neural networks are widely adopted in ASR, but their performance is highly dependent on architectural design

choices, particularly network depth [10], [11]. Since no principled rule exists to define the optimal number of layers, architectures are often selected through trial and error. This work addresses this limitation by proposing a collaborative layer-insertion strategy that incrementally defines network depth, adding parallel hidden-layer branches and retaining them only when measurable learning improvements are achieved.

## II. RELATED WORKS

Several studies address automated neural network design. Bayesian optimization has been applied to tune learning rate, depth, and activation functions, outperforming manual and grid search [12]. Comparisons with genetic algorithms highlight tradeoffs between exploration capability and computational cost [13]. Particle swarm optimization has also been used to design stacked autoencoders with competitive accuracy and reduced complexity [14], while optimized recurrent models show strong performance in speech recognition [15].

Constructive neural models originate from the CCNN framework. FCCN incrementally adds neurons while training only output weights, reducing training cost [16]. Other variants analytically compute weights [17], dynamically insert neurons [18], or integrate evolutionary strategies to obtain compact and stable architectures [19].

In speech recognition, Portuguese phonetic syllable classification using constructive stacked autoencoders was investigated in [20]. Classical MLP-based methods remain effective for small vocabularies [21], whereas recent models such as SincNet [22] and modular expert networks [23] achieve higher accuracy at the cost of increased computational complexity. In contrast to these approaches, the present work proposes a collaborative layer-insertion strategy that incrementally defines network depth through parallel branches, reducing architectural design effort without relying on extensive hyperparameter optimization.

### III. METHODOLOGY

The algorithms were implemented in Python using Google Colab. The experiments employed the *Male/Female for Forced Phonetic Alignment* dataset, which consists of clean studio recordings from one male and one female Brazilian Portuguese speaker and is originally highly unbalanced across phonetic classes [24]. To mitigate classification bias, avoid dominance of frequent classes, and allow a fair evaluation of the learning capacity of the proposed model, a balanced subset was constructed by selecting 43 phonetic syllable classes with 30 samples each, resulting in 1,290 syllable instances. Audio pre-processing and phonetic segmentation were performed using Praat, UFPAlign, and Kaldi, producing syllable-level signals used for feature extraction [25]–[27].

#### A. Collaborative layer insertion

The main novelty of this work is a collaborative layer-insertion strategy that incrementally defines network depth by adding parallel hidden-layer branches, which are retained only when they provide measurable learning improvement, avoiding exhaustive hyperparameter search. To mitigate error degradation and vanishing gradients [28], [29], new layers are inserted sequentially while preserving prior learning. As illustrated in Fig. 1, the initial network (solid black lines) is first trained and then frozen. A new branch is added using three types of connections: random intra-branch weights (blue dashed), inter-branch connections to previous layers (red dashed), and zero-initialized weights (black dashed).

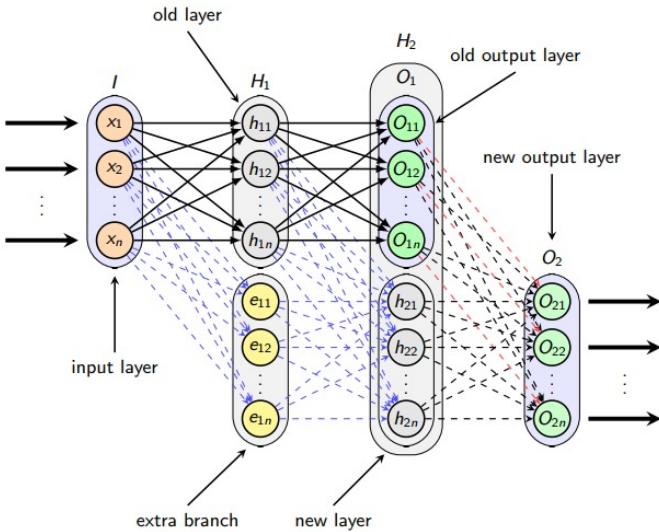


Fig. 1: Collaborative process of inserting a new hidden layer

The network input is  $X = (x_1, x_2, \dots, x_n)$ , where  $n$  is the size of the input and output layers. The vector  $H_1 = (h_{11}, h_{12}, \dots, h_{1m})$  represents the hidden-layer activation of branch 1 after applying  $f$  (Eq. 1), using weight matrix  $W_1^1$  and bias vector  $b_1^1$ . Eq. 2 defines the output layer.  $W_2^1$  are the weights from the hidden layer to the output layer, and  $b_2^1$

is the output bias. Branch 2 consists of a hidden layer and an output layer. Its input comes from the last hidden layer of Branch 1, and the output of Branch 1 is concatenated with the hidden layer of Branch 2. To expand the search space and reduce the overhead caused by adding new hidden layers, an extra branch can be added. This branch connects the input layer directly to the new hidden layer and is represented as  $E = (e_1, e_2, \dots, e_n)$ . Thus, the output of the hidden layer  $H_2$  of Branch 2 is defined by Eq. 3, where  $W_2^2$  is the weight matrix between the hidden and output layers.

$$H_1 = f(W_1^1 X + b_1^1) \quad (1)$$

$$O_1 = f(W_2^1 H_1 + b_2^1) \quad (2)$$

$$H_2 = [f(W_1^2 H_1 + W_2^e E + b_2^2) \quad O_1]^T \quad (3)$$

The output layer of the new branch has connections with the elements of its hidden layer ( $w_2^2$ ) and with the old branch through the weight matrix  $I_k(n)$ . Together, these components form the weight matrix  $W_2^2$ , defined in Eq. 4. The weights  $w_2^2$  are initialized to zero, ensuring that the new neurons do not affect the knowledge already learned by the network at the beginning of training. The matrix  $I_k(n)$  is diagonal, with its main diagonal elements set to the value  $k$ . The network output is provided by branch 2, defined by Eq. 5. Training branch 2 aims to compensate for the output of branch 1. Since the connections  $k$  have values equal to one and the activation function of the output layer is linear, the neural network will have only one output layer (branch 2 output layer), formed by combining the outputs of the two branches.

$$W_2^2 = [I_k(n) \quad w_2^2] = \begin{bmatrix} k & 0 & \cdots & 0 & 0 & 0 & \cdots & 0 \\ 0 & k & \cdots & 0 & 0 & 0 & \cdots & 0 \\ \vdots & \vdots & \ddots & \vdots & \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & \cdots & k & 0 & 0 & \cdots & 0 \end{bmatrix} \quad (4)$$

$$O_2 = f(W_2^2 H_2 + b_2^2) \quad (5)$$

#### B. Feature Extraction Using Centroids

After preprocessing [20], the FFT (first 8 kHz) converts audio to frequency rectangles. Adjacent frequencies form blocks summarized by centroids  $C_m = (X_m, Y_m)$  (Eq. 6, 7). Fig. 2 shows grouping: blue points are bar centers, red points are centroids. Sixteen blocks of 50 frequencies yield 16 centroids per syllable, normalized to  $[0, 1]$  for input.

$$X_m = \frac{\sum_{f=1}^n x_f \times a_f}{\sum_{f=1}^n a_f} \quad (6)$$

$$Y_m = \frac{\sum_{f=1}^n \frac{y_f}{2} \times a_f}{\sum_{f=1}^n a_f} \quad (7)$$

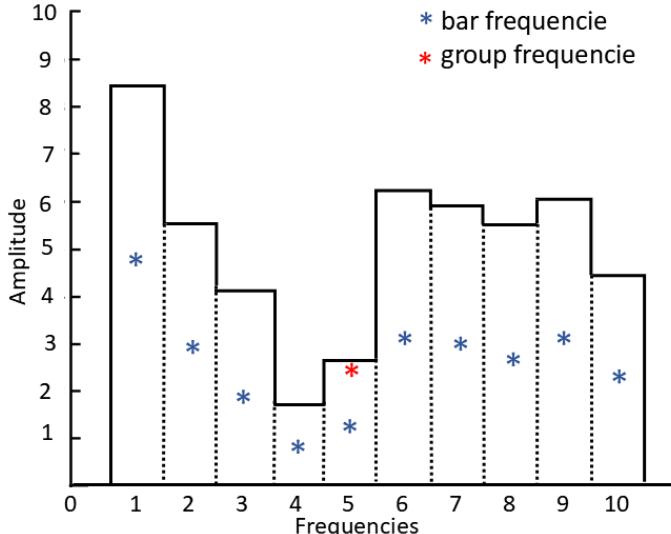


Fig. 2: Representation of a group of 10 frequency intervals.

### C. Evaluation of Experiments

MSE was used for training; accuracy, precision, sensitivity, and F1-score validated performance. Growth strategies were constant, increasing, decreasing, and random. Weight  $k$  variations: M1 – constant without extra branch; M2 – adjustable without extra branch; M3 – constant with extra branch; M4 – adjustable with extra branch.

Network depth was defined incrementally using the proposed collaborative insertion strategy. Four hidden layers were added sequentially, resulting in one input layer, five hidden layers, and one output layer. The input consisted of centroid coordinates  $(X_m, Y_m)$  (32 features), with 25 neurons in the first hidden layer (empirically defined) and 43 outputs. Learning rates were  $1 \times 10^{-5}$  (input-hidden) and  $5 \times 10^{-5}$  (hidden-output). Hidden layers used tanh, the output layer was linear, and performance was evaluated using 5-fold cross-validation.

## IV. RESULTS

The following figures analyze the effect of different layer-insertion strategies and collaboration mechanisms on training error and generalization. Fig. 3 shows errors for constant-type insertions; dashed black lines indicate the scenario without new layers, allowing direct comparison. Table I presents training, validation, and test MSE, along with the training error of each inserted layer. For the constant type, all methods achieved similar final errors. Extra depth reduced MSE in all methods (M1–M4), with the largest drops in the first three insertions, while still outperforming the baseline. Methods M2 and M4 achieved the lowest errors, especially in deeper networks.

Figure 4 shows the results of incremental insertion. The largest gains occur in the first added layers, with a clear error drop between layers 1 and 3. Methods M2 and M4 reach the lowest errors, especially in deeper configurations, indicating that adjusting the weight  $k$  improves learning. These results confirm that progressive layer insertion enhances modeling and that the branch-connection mechanism gives the new layers greater learning flexibility.

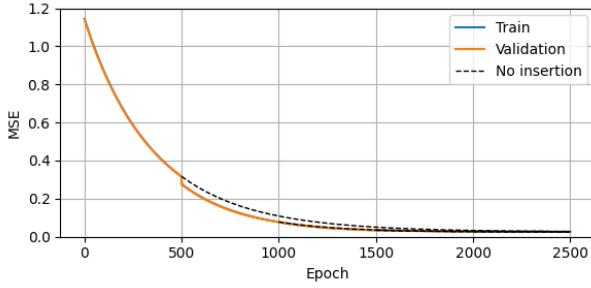
In the descending insertion method, Fig. 5, methods M2 and M4 achieve greater error reductions than M1 and M3. Table I shows that M2 and M4 consistently outperform the others for any depth, indicating greater robustness and generalizability.

In random insertion, each new layer has a dynamic number of neurons. Fig. 6 shows that M1 and M3 do not consistently reduce the training error, except in the third layer, while M2 and M4, with the outer branch and adjustable  $k$ , reduce the error. Table I shows that varying the number of neurons offers limited gains, but M2 and M4 achieve smaller errors (0.034 and 0.032) compared to M1 and M3 (0.066 and 0.058), reflecting better adaptation.

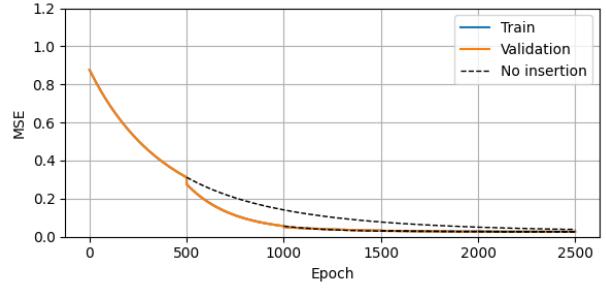
Table II compares the best results of this work (collaborative insertion) with the method proposed by [20]. The collaborative insertion model outperformed the literature, demonstrating consistent classification capabilities across all insertion types. In terms of accuracy, the proposed approach exceeded 76% with only two layers inserted, and the lowest value achieved was 87% for methods M1 and M3 under random insertion.

TABLE I: Mean squared error (MSE) obtained for each layer-insertion strategy.

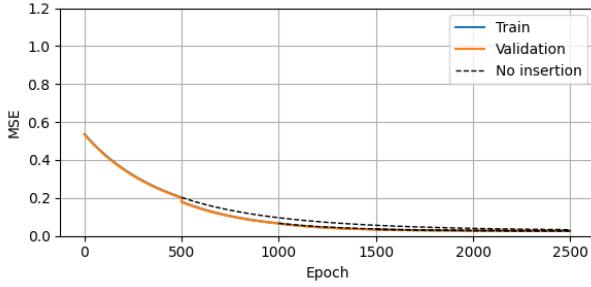
Type	Method	Layer 1			Layer 2			Layer 3			Layer 4			Layer 5		
		Train	Val.	Test												
Constant	M1	0.317	0.315	0.320	0.078	0.077	0.079	0.034	0.034	0.035	0.028	0.029	0.026	0.026	0.026	0.027
	M2	0.312	0.312	0.307	0.056	0.056	0.055	0.033	0.032	0.031	0.028	0.028	0.027	0.027	0.026	0.026
	M3	0.203	0.201	0.203	0.067	0.066	0.067	0.034	0.034	0.034	0.028	0.028	0.028	0.027	0.027	0.027
	M4	0.170	0.168	0.171	0.048	0.047	0.049	0.031	0.030	0.031	0.027	0.027	0.027	0.026	0.026	0.026
Increasing	M1	0.267	0.266	0.266	0.078	0.077	0.077	0.036	0.036	0.036	0.034	0.034	0.034	0.034	0.034	0.034
	M2	0.286	0.284	0.289	0.051	0.051	0.052	0.029	0.029	0.029	0.028	0.028	0.028	0.028	0.028	0.028
	M3	0.295	0.296	0.294	0.101	0.101	0.100	0.034	0.034	0.033	0.033	0.033	0.032	0.032	0.032	0.032
	M4	0.246	0.246	0.245	0.045	0.045	0.044	0.027	0.028	0.027	0.027	0.027	0.027	0.027	0.027	0.027
Decreasing	M1	0.324	0.321	0.325	0.132	0.130	0.132	0.065	0.064	0.065	0.042	0.042	0.042	0.037	0.037	0.037
	M2	0.270	0.272	0.266	0.064	0.065	0.062	0.038	0.039	0.037	0.032	0.033	0.031	0.030	0.030	0.029
	M3	0.352	0.350	0.346	0.135	0.134	0.131	0.064	0.063	0.061	0.046	0.045	0.043	0.038	0.038	0.037
	M4	0.322	0.319	0.321	0.054	0.053	0.054	0.033	0.033	0.033	0.030	0.029	0.029	0.029	0.028	0.028
Random	M1	0.250	0.248	0.249	0.180	0.178	0.179	0.101	0.103	0.103	0.085	0.084	0.084	0.066	0.065	0.065
	M2	0.270	0.271	0.269	0.090	0.091	0.090	0.042	0.043	0.042	0.038	0.038	0.038	0.034	0.034	0.034
	M3	0.296	0.300	0.298	0.205	0.208	0.207	0.099	0.101	0.101	0.085	0.087	0.087	0.058	0.059	0.059
	M4	0.309	0.308	0.310	0.082	0.082	0.083	0.041	0.041	0.041	0.037	0.037	0.037	0.032	0.032	0.032



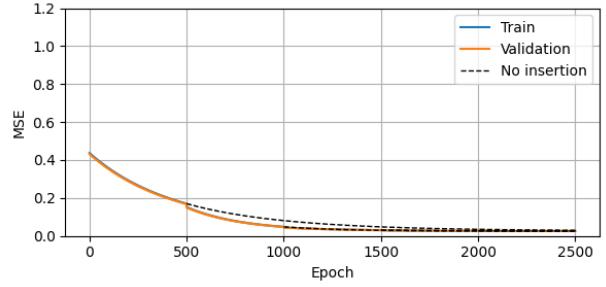
(a) Method M1



(b) Method M2

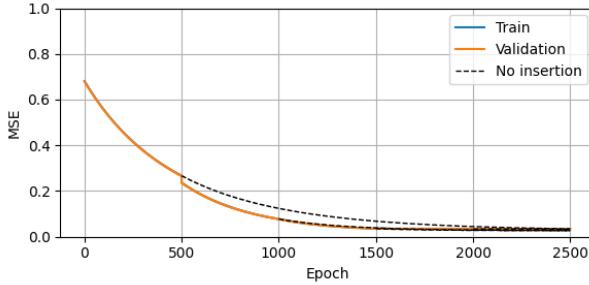


(c) Method M3

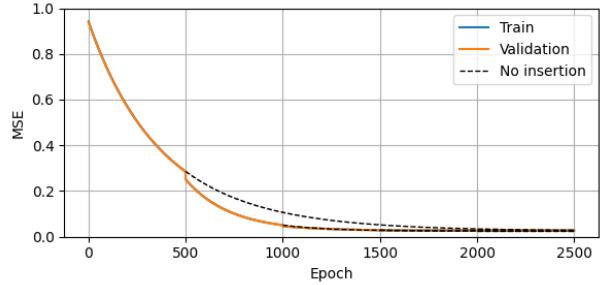


(d) Method M4

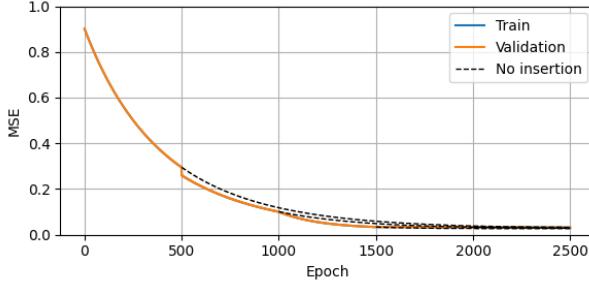
Fig. 3: Mean squared error (MSE) across successive constant layer insertions.



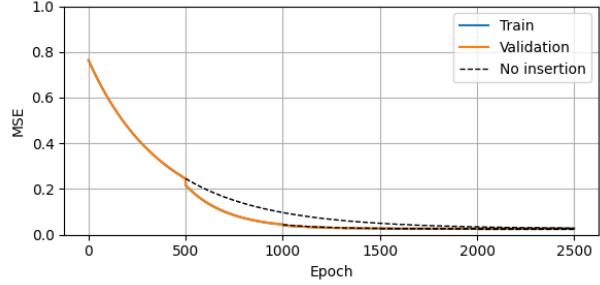
(a) Method M1



(b) Method M2



(c) Method M3



(d) Method M4

Fig. 4: Mean squared error (MSE) across successive increasing layer insertion.

TABLE II: Performance comparison with related works.

Metric	Collabnet	Collaborative insertion
accuracy	0.76	0.98
sensitivity	0.76	0.89
precision	0.82	0.93
<i>f1-score</i>	0.77	0.91

## V. CONCLUSION

The insertion types and methods show that a collaborative branching strategy for adding hidden layers to stacked autoencoders improves performance, overfitting control, and reduces hyperparameter tuning effort, while enhancing scalability and

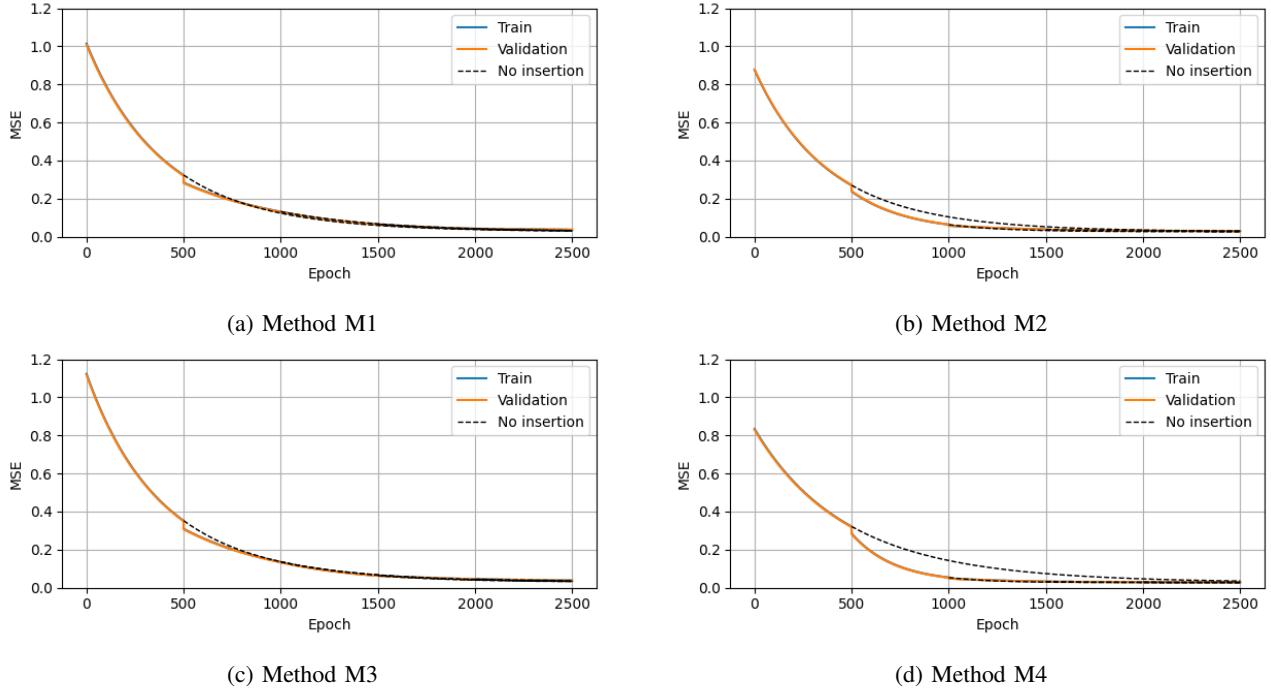


Fig. 5: Mean squared error (MSE) across successive decreasing layer insertion.

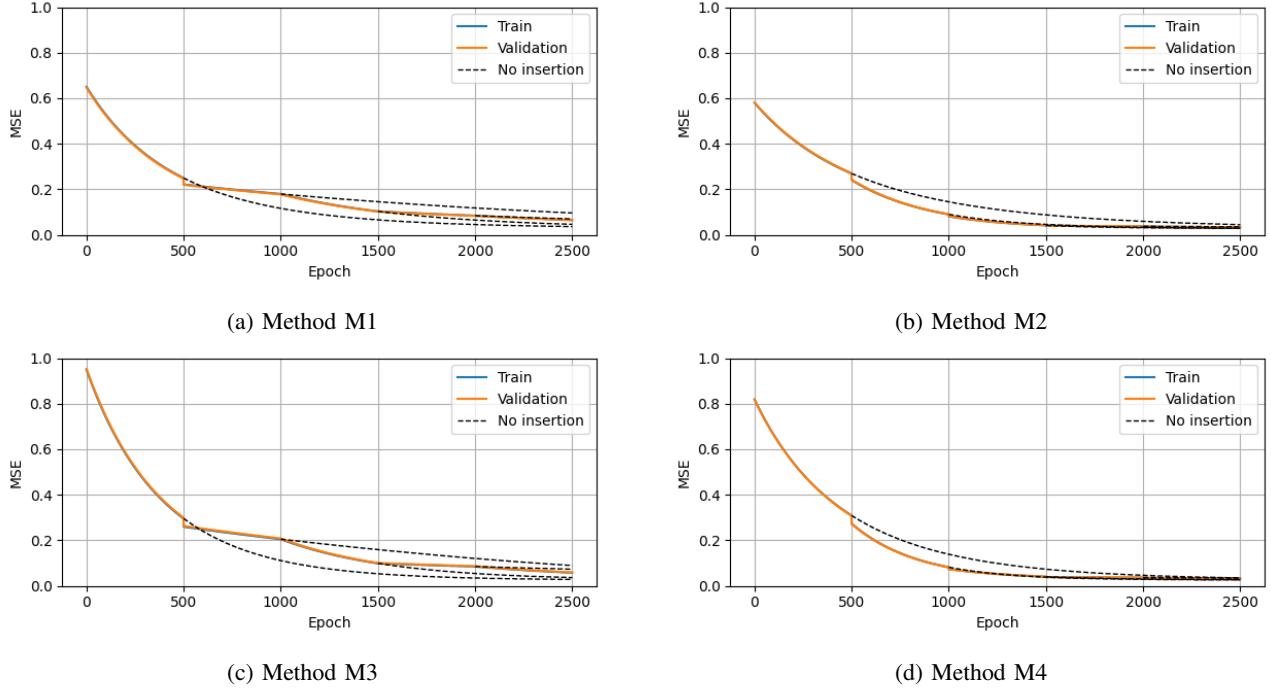


Fig. 6: Mean squared error (MSE) across successive random layer insertion.

design efficiency.

The centroid-based representation efficiently compresses phonetic-syllable data, reducing dimensionality and computational cost while preserving relevant spectral information for learning. Although the experiments used a controlled dataset with two speakers, the results validate the approach

as a proof of concept for adaptive depth construction in stacked autoencoders. Future work includes evaluation on larger datasets, comparison with modern speech recognition architectures, and investigation of dynamic activation-function switching after layer insertion to further enhance adaptability.

and performance.

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