Intrinsic Noise Improves Speech Recognition in a Computational Model of the Auditory Pathway

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Noise is generally considered to harm information processing performance. However, in the context of stochastic resonance, noise has been shown to improve signal detection of weak sub-threshold signals, and it has been proposed that the brain might actively exploit this phenomenon. Especially within the auditory system, recent studies suggest that intrinsic noise plays a key role in signal processing and might even correspond to increased spontaneous neuronal firing rates observed in early processing stages of the auditory brain stem and cortex after hearing loss. Here we present a computational model of the auditory pathway based on a deep neural network, trained on speech recognition. We simulate different levels of hearing loss and investigate the effect of intrinsic noise. Remarkably, speech recognition after hearing loss actually improves with additional intrinsic noise. This surprising result indicates that intrinsic noise might not only play a crucial role in human auditory processing, but might even be beneficial for contemporary machine learning approaches.

Keywords: speech processing, auditory perception, hearing loss, stochastic resonance, deep artificial neural networks, dorsal cochlear nucleus, tinnitus mechanisms, Zwicker tone

INTRODUCTION

The term noise usually describes undesirable disturbances or fluctuations, and is considered to be the “fundamental enemy” (McDonnell and Abbott, 2009) for communication and error-free information transmission and processing in engineering. However, a vast and still increasing number of publications demonstrate the various benefits of noise for signal detection and processing, among which the most important phenomena are called stochastic resonance (McDonnell and Abbott, 2009), coherence resonance (Pikovsky and Kurths, 1997), and recurrence resonance (Krauss et al., 2019a).

The term stochastic resonance (SR), first introduced by Benzi et al. (1981), refers to a processing principle in which signals that would otherwise be sub-threshold for a given sensor can be detected by adding a random signal of appropriate intensity to the sensor input (Benzi et al., 1981; Gammaitoni et al., 1998; Moss et al., 2004). SR occurs ubiquitously in nature and covers a broad...
inteactive role in a vast number of different systems (Douglass et al., 1993; Collins et al., 1996; Gluckman et al., 1996; Nozaki et al., 1999; Usher and Feingold, 2000; Ward et al., 2002; Kosko and Mitaim, 2003; Aihara et al., 2008; Faisal et al., 2008). Also, it has already been proposed that spontaneous random activity, i.e., noise, may increase information transmission via SR in the auditory brain stem (Mino, 2014).

In self-adaptive signal detection systems based on SR, the optimal noise intensity is continuously adjusted via a feedback loop so that the system response remains optimal in terms of information throughput, even if the characteristics and statistics of the input signal change. The term adaptive SR was coined for this processing principle (Mitaim and Kosko, 1998, 2004; Wenning and Obermayer, 2003). In a previous study we demonstrated that the auto-correlation of the sensor output, a quantity always accessible and easy to analyze by neural networks, can be used to quantify and hence maximize information transmission even for unknown and variable input signals (Krauss et al., 2017).

In further studies we demonstrated theoretically and empirically that adaptive SR based on output auto-correlations might be a major processing principle of the auditory system that serves to partially compensate for acute or chronic hearing loss, e.g., due to cochlear damage (Krauss et al., 2016, 2018, 2019b; Gollinast et al., 2017; Krauss and Tziridis, 2021; Schilling et al., 2021d). Here, the noise required for SR would correspond to increased spontaneous neuronal firing rates in early processing stages of the auditory brain stem and cortex, and would be perceived as a phantom perception. Remarkably, this phenomenon has frequently been observed in animal models and in humans with subjective tinnitus (Wang et al., 1997; Ahlf et al., 2012; Tziridis et al., 2015; Wu et al., 2016), which in turn is assumed to be virtually always caused by some kind of apparent hearing loss (Krauss et al., 2006; Shore and Zhou, 2006; Koehler and Shore, 2013; Wu et al., 2016; Ansorge et al., 2021; Niven and Scott, 2021), and that noise trauma alters long-term somatosensory-auditory processing in the DCN (Dehmel et al., 2012), i.e., somatosensory projections are up-regulated after deafness (Zeng et al., 2012).

In self-adaptive signal detection systems based on SR, the optimal noise level is continuously adjusted so that the system response in terms of information throughput remains optimal, even if the properties of the input signal change. The term adaptive SR was coined for this processing principle (Mitaim and Kosko, 1998, 2004). An objective function for quantifying information content is the mutual information between the sensor input and the output (Shannon, 1948), which is often used in theoretical approaches (Levin and Miller, 1996; Mitaim and Kosko, 2004; Moss et al., 2004). The choice of mutual information is obvious, since the basic purpose of each sensor is to transmit information to a subsequent information processing system. It has already been shown that the mutual information has a maximum as a function of the noise intensity, which indicates the optimal noise level that has to be added to the input signal in order to achieve optimal information transmission by SR (Moss et al., 2004). A fundamental disadvantage of the mutual information, however, is the impossibility of calculating it in every application of adaptive SR if the signal to be recognized is unknown (Krauss et al., 2017). Even if the underlying signal is known, the use of mutual information in the context of neural network architectures seems to be rather impractical, since its calculation requires the evaluation of probability distributions, logarithms, products and fractions, i.e., operations difficult to implement in neural networks. In an earlier work (Krauss et al., 2017) we were able to show that this fundamental disadvantage can be overcome by another objective function, namely the autocorrelation of the sensor response. Both, the mutual information and the autocorrelation peak at the same noise level. Hence, maximization of the output autocorrelation leads to similar or even identical estimates of the optimal noise intensities for SR as the mutual information, but with the decisive advantage that no knowledge of the input
signal is required (Krauss et al., 2017). In contrast to mutual information, the evaluation of autocorrelation functions in neural networks can easily be implemented using delay lines and coincidence detectors (Licklider, 1951). Remarkably, a cerebellar-like neuronal architecture resembling such delay-lines is known to exist in the DCN (Osen et al., 1988; Hackney et al., 1990; Nelken and Young, 1994; Oertel and Young, 2004; Baizer et al., 2012). Therefore, we previously proposed the possibility that the neural noise for SR is injected into the auditory system via somatosensory projections to the DCN (Krauss et al., 2016, 2018, 2019b; Krauss and Tziridis, 2021; Schilling et al., 2021d, 2022). The idea that central noise plays a key role in auditory processing has recently gained increasing popularity (Zeng, 2013, 2020; Koops and Eggermont, 2021) and is supported by various findings. For instance, it is well-known, that jaw movements lead to a modulation of subjective tinnitus loudness (Pinchoff et al., 1998). This may easily be explained within our framework, as jaw movements alter somatosensory input to the DCN. Since this somatosensory input corresponds to the noise required for SR, auditory input to the DCN is modulated through this mechanism, and the altered noise level is then perceived as modulated tinnitus (Krauss et al., 2016, 2018, 2019b; Schilling et al., 2021d). Along the same line, one may explain why both, the temporomandibular joint syndrome and whiplash, frequently cause so called somatic tinnitus (Levine, 1999). Another example is the finding of Tang and Trussell (2015, 2017), who demonstrated that somatosensory input and hence tinnitus sensation may also be modified by serotonergic regulation of excitability of principal cells in the DCN. In addition, DCN responses to somatosensory stimulation are enhanced after noise-induced hearing loss (Shore et al., 2008; Shore, 2011). Finally, and most remarkable, electro-tactile stimulation of finger tips, i.e., increased somatosensory input, significantly improves both, melody recognition (Huang et al., 2019) and speech recognition (Huang et al., 2017) in patients with cochlear implants.

However, while we propose the DCN to be the place where auditory input from the cochlea is integrated with neural noise from the somatosensory system, we cannot rule out that SR rather occurs in the ventral cochlear nucleus (VCN) instead (see “Discussion” section).

In order to further support the hypothesis that SR plays a key role in auditory processing, we here present a hybrid computational model of the auditory pathway, trained on speech recognition. An overview of the model layout is provided in Figure 1.

In our approach, also the DCN circuitry is not modeled in all detail, but only as a one-layered structure of leaky-integrate-and-fire neurons, which are not interconnected. The aim of our implementation is not to understand the whole auditory pathway in detail, which would be far to ambitious, but to find out if SR could have a significant effect on speech perception. Thus, it is not the aim of the study to analyze the auditory system on an implementational level (see Marr’s level of analysis; Marr and Poggio, 1979), but to explain the algorithmic level (Krauss and Schilling, 2020; Schilling et al., 2022).

The output of the DCN is fed to a deep neural network trained on word recognition. The deep neural network can be interpreted as a surrogate for all remaining stages of the auditory pathway beyond the DCN up to the auditory cortex. However, it may also be regarded as a tool to quantify the information content of the DCN output. The deep neural network was trained once on a training data set and kept stable for the experiments.

Furthermore, we simulate different levels of hearing loss (cochlear damage) and compare the resulting word recognition accuracies for with the accuracy of the non-disturbed model (i.e., without simulated hearing loss). Subsequently, we add intrinsic noise of different intensities to the model. The overall data flow in our model is depicted in Figure 2.

As expected and shown in various experimental studies with human subjects (Lorenzi et al., 2006; Zeng and Liu, 2006) we find in our model that speech recognition accuracy decreases systematically with increasing hearing loss (Zeng and Djalilian, 2010). In the case of additional intrinsic noise, we find SR-like behavior for all levels of hearing loss: depending on the intensity of the noise, accuracy first increases, reaches a peak, and finally decreases again. This means that speech recognition after hearing loss may indeed be improved by our proposed mechanism. A simple increase of the spontaneous activity of the DCN neurons did not lead to an increased speech recognition, which indicates that indeed SR causes the increase in word recognition accuracy. This intriguing result indicates, that SR indeed plays a crucial role in auditory processing, and might even be beneficial for contemporary machine learning approaches.

RESULTS
Dorsal Cochlear Nucleus Model Neurons Show Phase Coupling Below 4 kHz
In order to validate our DCN model, we investigate the spike train output of the 30 leaky integrate- and-fire (LIF) neurons for different sine wave inputs (Figure 3). As described in detail in “Methods” section, the parameters of the LIF neurons are chosen so that the refractory time (0.25 ms) of the neurons does not allow for firing rates above 4 kHz. This is much more than the maximum spiking rate of a biological...
FIGURE 1 | Model layout. The complete model consists of three different modules representing different stages of the auditory pathway in the human brain. The input to the model are single words encoded as wave files with a sampling rate of 44.1 kHz and 1 s duration (A). The cochlea and the spiral ganglion are modeled as an array of 30 band-pass filters (B). The continuous output signal of (B) serves as input to 30 leaky integrate-and-fire-neurons representing the DCN (C). The spike-train output of the DCN model is down sampled and serves as input for a deep neural network that is trained with error backpropagation on the classification of 207 different German words (D). The classification accuracy serves as a proxy for speech recognition (E). In order to investigate the effect of a particular hearing loss, the cochlea output amplitude is decreased by a certain factor independently for all frequency channels (F). White noise representing somatosensory input to the DCN can be added independently to the input of the different leaky-integrate-and-fire-neurons (LIF, G).

FIGURE 2 | Data flow in auditory pathway model. The scheme shows how the speech data is processed within the model. The cochlea splits the signal via 30 bandpass filters. The bandpass filtered data is scaled down to simulate a hearing loss. The hearing loss affects only channels within the speech relevant frequency range (orange, green, red). The other frequency channels are unchanged. Neural noise is added to investigate the effect of stochastic resonance (only in hearing impaired channels). The DCN is simulated as 30 LIF neurons. Each LIF neuron represents a complete biological neuron population. The spike data is down-sampled and fed to the deep neural network.

neuron (400 Hz) (Nizami, 2002). However, the recruitment of several neurons to increase the frequency range in which phase coupling is possible is a core concept within the dorsal cochlear nucleus (Langner, 1988). Thus, in our model 1 simulated LIF neuron represents approximately 10 biological neurons, having individual refractory times above 1 ms.

We find that for stimulus frequencies above 4 kHz and amplitudes of 0.001 the LIF neurons do not spike at all (Figure 3A). In contrast, for a larger amplitude of 0.002, a rate coding without phase coupling can be observed (Figure 3B). Furthermore, we find that the LIF neurons are sensitive to amplitude modulations also in the frequency range above 4 kHz (Figures 3C,D). Thus, our DCN neurons are designed so that they allow for phase coupling in the frequency range crucial for speech comprehension, as is known from the human auditory system.

Word Processing From Cochlea to Dorsal Cochlear Nucleus

In analogy to the auditory system, the complex auditory stimuli representing spoken words (Figure 4A) are transformed in the cochlea into continuous signals in a number of different frequency channels, in our model 30. However, the cochlea does
not perform a simple Fourier transform, but rather splits the signal into multiple band pass filtered signals, thereby preserving the complete phase information (Figure 4B). For the purpose of simplicity, in the context of our model we assume that the auditory nerve fibers directly transmit this analog signal to the DCN, which is regarded to be a special feature of the auditory system (Kandel et al., 2000; Young and Davis, 2002).

The analog signals are then further transformed into spike train patterns in the DCN (Figure 4C). Thus, each spoken word is represented as a unique spiking pattern with a dimensionality...
FIGURE 4 | Exemplary processing of a word in cochlea and DCN model. (A) The first 0.2 s of audio data of the German word “die” (the). (B) The 30 frequency components (blue without hearing loss, orange with hearing loss) after the first part of the model, which represents the cochlea and the spiral-ganglion (Figure 1A). A virtual hearing loss is applied by weakening the signal at a certain frequency range (e.g., 400 Hz–4 kHz, −30 dB). The bandpass filtered signal (matrix of 30 frequency channels and $f_s \times$ signal duration) is fed to the LIF neurons (refractory time: $\approx 0.25$ ms) and spike trains (C) are generated. These spike trains are down-sampled by a factor of 5 and fed to the deep neural network (D). (E) The same signal (of panel D) with added hearing loss of 30 dB in the frequency range 400 Hz–4 kHz being the speech relevant range.

Note that we down-sampled these matrices by a factor of five from 44,100 to 8,200 Hz for deep learning (Figure 4D). This does not affect the phase coupling information in the speech relevant frequency range. In order to analyze speech processing in an impaired auditory system, we simulated a hearing loss in the speech relevant frequency range (400 Hz–4 kHz) by decreasing...
the cochlea output amplitudes by a certain factor. The weakened cochlea outputs and the resulting modified DCN spike train outputs are shown in Figures 4B,C, where orange corresponds to an exemplary hearing loss of 30 dB, and blue corresponds to the undisturbed signals, i.e., without hearing loss. The corresponding down sampled spike pattern matrices used as test data for the deep neural network, are shown in Figure 4D (without hearing loss) and in Figure 4E (with 30 dB hearing loss). We provide an exemplary overview of the effect of different hearing losses from 0 to 45 dB on the spike pattern matrices in Figure 5.

**Intrinsic Noise Partially Restores Spike Patterns After Simulated Hearing Loss**

To test the putative beneficial effect of intrinsic noise in case of hearing loss, we analyzed spiking patterns generated with and without intrinsic noise and compared them with the corresponding undisturbed patterns (Figure 6). In Figure 6A a sample spike pattern in case of no hearing loss is shown as reference. As expected, a simulated hearing loss of 30 dB in the frequency range of 400 Hz to 4 kHz leads to a decreased spiking activity (Figure 6B), which can be partially restored by the addition of intrinsic noise with optimal intensity (Figure 6C).

A point-to-point comparison of the spikes resulting from the undisturbed system (no hearing loss) with the spikes resulting from hearing loss and additional intrinsic noise, demonstrates that there is indeed some improvement. In Figure 6D only those spikes are shown that occur in both mentioned cases. In contrast, there are less spikes resulting from hearing loss without intrinsic noise (Figure 6E). Further analysis yield that intrinsic noise not only restores spikes correctly (Figure 6F, yellow), but also introduces false positive spikes (Figure 6F, blue). However, a direct point-to-point comparison of spike patterns does not fully capture the benefit of intrinsic noise. As shown in Figures 6A,C (green boxes), intrinsic noise even restores larger spatio-temporal spiking patterns correctly, yet with some temporal shift.

**Intrinsic Noise Improves Accuracy for Speech Recognition After Simulated Hearing Loss**

We also analyzed the effect of intrinsic noise on speech recognition accuracy in case of hearing loss in different scenarios.

![Figure 5](https://example.com/fig5.png) Compressed spike patterns with added hearing loss. The figure shows the down-sampled spike patterns of the same word as shown in Figure 4. The speech relevant frequency range (400 Hz–4 kHz) is artificially weakened (hearing loss). Panels (A–J) refer to hearing losses 0–45 dB.
Using our custom-made data set, we investigated hearing loss in two different frequency ranges. Furthermore, using the free spoken digit data set (FSDD) data set, we investigated hearing loss using two different neural networks. In all cases, we find that intrinsic noise of appropriate intensity improves accuracy for speech recognition after simulated hearing loss. Note that the weights of the deep neural network are kept constant for all further analyses. Thus, the relative accuracy is normalized to the original test accuracy (0.37) of the undisturbed network.

Custom-Made Data Set and Hearing Loss in the Frequency Range of 400 Hz–4 kHz

For the first scenario, we used a convolutional neural network (Table 1) trained on our custom-made data set. After training, we simulated a hearing loss in the frequency range of 400 Hz to 4 kHz which is known to be crucial for speech comprehension in humans (Fox, 2006). The effect of improved or decreased speech comprehension is quantified by the classification accuracy of the words (test accuracy). The classification accuracy as a function of the hearing loss has a biologically plausible sigmoid shape (Figure 7A dark blue curve). The test accuracies as a function of the added noise for different hearing losses show a clear resonance curve with a global maximum (Figure 7B). For a hearing loss of about 20 dB, the relative improvement of speech comprehension is more than doubled (Figure 7C). Furthermore, it can be shown that the optimal noise level correlates with the hearing loss (Figure 7D). This effect is plausible as for a weaker signal a higher noise amplitude is needed to lift the signal over the threshold of the LIF neurons. In summary, it can be stated that the addition of noise can lead to an improved speech comprehension for all hearing losses. This fact can be seen in Figure 7A, where the cyan curve shows the test accuracy as a function of the hearing loss with the ideal amount of added Gaussian noise.

### Custom-Made Data Set and Hearing Loss in the Frequency Range Above 4 kHz

Since many people suffer from hearing losses in the high frequency range (Ciorba et al., 2011). In the next step, the stochastic resonance effect is analyzed for a high frequency range.

#### Table 1 | Exact parameters of the used deep convolutional network (main analysis).

| Layer | Type | Input-output-dim | Activation | Characteristics |
|-------|------|-----------------|------------|------------------|
| 1     | Convolution 2D | 30, 8,820, 1;1, 8,791, 128 | ReLu       |                  |
| 2     | Reshape       | 1, 8,791, 128; 8,791, 128 |            |                  |
| 3     | Convolution 1D | 8,791, 128, 8,782, 128 | ReLu       |                  |
| 4     | MaxPooling 1D | 8,782, 128; 4,391, 128 | Pool size: 2 |                  |
| 5     | Dropout       | 4,391, 128, 4,391, 128 | Dropout: 0.5 |                  |
| 6     | Convolution 1D | 4,391, 128, 4,391, 128 | ReLu       |                  |
| 7     | Convolution 1D | 4,391, 128, 4,390, 128 | ReLu       |                  |
| 8     | MaxPooling 1D | 4,390, 128, 2,195, 128 | Pool size: 2 |                  |
| 9     | Dropout       | 2,195, 128, 2,195, 128 | Dropout: 0.5 |                  |
| 10    | Flatten       | 2,195, 128, 280,960 |            |                  |
| 11    | Dense         | 280,960; 150 | ReLu       |                  |
| 12    | Dense         | 150; 50 | ReLu       |                  |
| 13    | Dense         | 50; 207 | Softmax    |                  |

### Figure 6 | Effect of intrinsic noise on the DCN output patterns.

(A) Spiking without HL (same as in Figure 5A). (B) Spiking with a HL of 30 dB (same as Figure 5G). (C) Spiking activity with HL and intrinsic noise of optimal intensity. Additional white noise increases spiking activity. (D) Point-to-point comparison of spiking patterns for no HL and with HL and intrinsic noise. Shown are only spikes that occur in both cases, i.e., that are not affected by HL or that are correctly restored by noise. (E) Point-to-point comparison of spiking patterns for no HL with HL and without intrinsic noise. Shown are only spikes that occur in both cases, i.e., that are not affected by HL. (F) Intrinsic noise of optimal intensity not only restores spikes correctly (yellow), but also introduces false positive spikes (dark blue). Intrinsic noise restores spatio-temporal spiking patterns correctly, yet with some temporal shift (green boxes in panels A,C, zoom of spike pattern in green box).
FIGURE 7 | Effect of SR on speech recognition. (A) The curve shows the relative accuracy of the trained neural network as a function of the hearing loss (red dashed line: chance level; \( \frac{1}{207} \)). The hearing loss (5–50 dB, 5 dB steps, frequency range of HL: 400 Hz–4,000 Hz) was implemented in the test data set and propagated through the pre-trained network. Thus, the cochlea output was multiplied with an attenuation factor (10\(^{-\frac{HL}{20}}\)). This output was then transformed using the integrate-and-fire neurons and fed in the neural network. (B) Relative accuracy as a function of the applied noise level for different hearing losses. Resonance curves with one global maximum at a certain noise level > 0 could be shown. (C) Best relative improvement as a function of the hearing loss. (D) Optimal noise level as a function of the hearing loss.

hearing loss starting at a frequency of 4 kHz. It can be shown that the high frequency loss does not affect the speech comprehension abilities in the same manner as hearing losses in the critical frequency range between 400 Hz and 4 kHz (Figure 8A). The relative accuracy does not drop below a value of 50%. Thus, the effect of stochastic resonance is also reduced (Figure 8B), which means a maximal relative improvement of approximately 10% (Figures 8C,D). Furthermore, there is no real resonance curve with one maximum at a certain noise frequency but a second maximum at a higher noise level (Figure 8B). To put it in a nutshell, we can state that the addition of noise can lead to a significant improvement of speech recognition.

Custom-Made Data Set and Hearing Loss With Non-linearity in the Frequency Range of 400 Hz to 4 kHz

So far, we simulated linear hearing loss in the model cochlea. However, it is known that different damages to the cochlea or the synapses from the cochlea to the cochlear nuclei yield to different degrees of non-linearity in hearing loss. Therefore, we also tested our model with an additional threshold of −50 dB, i.e., all values above \(-10^{\frac{-50}{20}}\) and below \(10^{\frac{-50}{20}} \approx 0.003\) are set to zero. Also in the case of an additional hard threshold, leading to real information loss, the SR effect still works. The added noise leads to a signal enhancement. Thus, the signal causes more spiking in the DCN (Figure 9A). Consequently, the relative speech recognition accuracy is partly restored by SR (Figure 9B).

Free Spoken Digit Data Set Data Set and Hearing Loss in the Frequency Range Above 400 Hz

In order to further demonstrate that the reported results are not limited to a certain data set, natural language or neural network architecture, we repeated our analyses using two further neural networks, an alternative convolutional neural network architecture (Table 2) and a network with Long-Short-Term-Memories (Table 3), both trained and tested on English language, i.e., the FSDD data set (Figure 10). A hearing loss in the critical frequency range for speech comprehension leads to a decrease in the classification accuracy (10a for the convolutional network and 10c for the Long-Short-Term-Memory network). Furthermore, the stochastic resonance effect in terms of a clear resonance curve with one maximum can be observed (Figures 10B,D).

METHODS

Computational Resources

The simulations were run on a desktop computer equipped with an i9 extreme processor (Intel) with 10 calculation cores.
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FIGURE 8 | Effect of SR on speech recognition (high frequency hearing loss). Same analysis as shown in Figure 7 for high frequency hearing loss. (A) The plots show the relative accuracy of the trained neural network as a function of the hearing loss (red dashed line: chance level). The high frequency hearing loss lead to different effects (10–50 dB, 10 dB steps, frequency range of HL: above 4,000 Hz). (B) The relative accuracy as a function of the noise has no clear maximum above the value for no added noise (nearly no SR). Furthermore, a second local maximum occurs. (C) The best relative improvement does not significantly increase over 10%. (D) Optimal noise level as a function of hearing loss shows similar behavior as for the hearing loss in the speech relevant frequency range (cf. Figure 7).

FIGURE 9 | SR effect with additional threshold. (A) Example of signal after simulated cochlea for one frequency channel and an additional threshold of −50 dB (−50 dB means that all values above −10 dB and below 10 dB ≈ 0.003 are set to zero). This threshold is introduced to show that the SR effect also works when hearing loss leads to a real information loss. The noise leads to a signal enhancement (cyan curve). Thus, the signal causes more spiking in the DCN (cyan dots compared to brown dots). (B) Relative accuracy as a function of the amplitude of the added noise. SR resonance partly restores the accuracy. For very high thresholds, where main parts of the signals are deleted, the SR does not restore the accuracy.

Furthermore, the machine learning was run on the same computer on two Nvidia Titan XP graphical processor units. To test the validity of our calculations the simulations were performed on two different code bases. The main results based on our own speech data set are mainly based on Numpy (Walt et al., 2011) and SciPy (Jones et al., 2001) calculations. The convolutional network was implemented in Keras (Chollet, 2018) with Tensorflow (Abadi et al., 2016) back-end. All main results were confirmed by analyzing a standard speech data set—the so called Jakobovski free spoken digit data set (FSDD) (Jackson et al., 2018), containing spoken numbers from 0 to 9 in English language in accordance to the MNIST data set with
TABLE 2 | Exact parameters of the used deep convolutional network (FSDD data set).

| Layer | Type            | Input-output-dim | Activation | Characteristics |
|-------|-----------------|------------------|------------|-----------------|
| 1     | Convolution 2D  | 9,131, 30, 1; 9,102, 1, 32 | ReLu       |                 |
| 2     | MaxPooling 2D   | 9,102, 1, 32; 4,551, 1, 32 | Pool size: (2, 1) |                 |
| 3     | DropOut         | 4,551, 1, 32; 4,551, 1, 32 | Dropout: 0.2 |                 |
| 4     | Convolution 2D  | 4,551, 1, 32; 4,547, 1, 64 | ReLu       |                 |
| 5     | MaxPooling 2D   | 4,547, 1, 64; 2,273, 1, 64 | Pool size: (2, 1) |                 |
| 6     | DropOut         | 2,273, 1, 64; 2,273, 1, 64 | Dropout: 0.2 |                 |
| 7     | Convolution 2D  | 2,273, 1, 64; 2,272, 1, 32 | ReLu       |                 |
| 8     | MaxPooling 2D   | 2,272, 1, 32; 1,136, 1, 32 | Pool size: (2, 1) |                 |
| 9     | DropOut         | 1,136, 1, 32; 1,136, 1, 32 | Dropout: 0.2 |                 |
| 10    | Flatten         | 1,136, 1, 32; 36,352 |            |                 |
| 11    | Dense           | 36,352; 400       | ReLu       |                 |
| 12    | DropOut         | 400; 400          | Dropout: 0.2 |                 |
| 13    | Dense           | 400; 50           | ReLu       |                 |
| 14    | Dense           | 50; 10            | Softmax    |                 |

TABLE 3 | Exact parameters of the used LSTM network (FSDD data set).

| Layer | Type            | Input-output-dim | Activation | Characteristics |
|-------|-----------------|------------------|------------|-----------------|
| 1     | GroupToBatches  | 9,000, 30; 45, 6,000 |            |                 |
| 2     | LSTM            | 45, 6,000; 45, 200; 45, 200 | tanh      |                 |
| 3     | DropOut         | 45, 200; 45, 200 | Dropout: 0.5 |                 |
| 4     | LSTM            | 45, 200; 45, 100 | tanh      |                 |
| 5     | DropOut         | 45, 100; 45, 100 | Dropout: 0.5 |                 |
| 6     | TimeDistributed | 45, 100; 45, 100 |            |                 |
| 7     | Dense           | 45, 100; 45, 100 | Dropout: 0.5 |                 |
| 8     | TimeDistributed | 45, 100; 45, 10  |            |                 |

written digits in this range (LeCun et al., 1998). This was done using a completely new code base exclusively build of KERAS layers. Thus, a custom-made KERAS layer implemented as sinc FIR filters for the cochlea layer as well as the leaky-integrate-and-fire neurons were implemented. All plots were created using the Matplotlib Python library (Hunter, 2007) and plots were arranged using the pylustrator (Gerum, 2020).

Layout of the Computational Model and General Approach

The model comprises three modules (Figure 1): (1) an artificial cochlea modeled as an array of band-pass filters, (2) a model of the dorsal cochlear nucleus (DCN), implemented as an array of leaky integrate-and-fire (LIF) neurons, and (3) a deep neural network, that represents all further processing stages beyond the DCN up to the auditory cortex and higher, language associated, cortex areas.

The input to the model are single words of spoken language encoded as wave files with a sampling rate of 44.1 kHz and 1 s duration (Figure 1A). These wave files represent the acoustic input of speech to the auditory system, and are processed in the first module of the model representing the cochlea and the spiral ganglion (Figure 1B). Like in previously published models (Moore and Glasberg, 1983; Houser et al., 2001; Sayles and Winter, 2010), this module is implemented as an array of rectangular band-pass filters. In order to limit the total computation time, we restricted our model to 30 band-pass filters, instead of the actual amount of approximately 3,500 inner hair cells in the human cochlea (Nadol, 1988). According to the physiology of the cochlea (Russell and Nilsen, 1997), the center frequencies of the band-pass filters are chosen such that they cover the frequency range from 100 Hz to 10 kHz in logarithmic steps.

The continuous multi-channel output of the band-pass filter array serves as input to an array of 30 LIF neurons (Burkitt, 2006) representing the DCN (Figure 1C). We here applied a one-to-one mapping from band-pass filters to model neurons, i.e., we do not explicitly account for putative cross-talk between neighboring frequency channels. However, since both the cochlea and the DCN model only consist of 30 different frequency channels, each of these channels may be regarded as an already coarse grained version of approximately 100 different frequency channels that exist in the human auditory system. Thus, eventual cross-talk is implicitly implemented in our model within each of the 30 modeled channels. The output of our DCN model comprises the spike trains of the 30 LIF neurons. Note that, in our DCN model, a single LIF neuron represents approximately 10 biological neurons processing the same frequency channel (Kandel et al., 2000).

In our cochlea and DCN model, the outputs of the band-pass filters and the membrane potentials of the LIF neurons change with the same rate (44.1 kHz) as the wave file input. However, the LIF neurons spike at lower average rates, due to their refractory period. It is therefore possible to down-sample this sparse output spike train, thereby reducing the data volume for the subsequent deep neural network. In order to preserve enough temporal information for phase coding, we down-sample the DCN output only by a factor of five, so that the 44,100 momentary amplitudes of the input wave file per second are finally transformed into a binary 30 × 8,820 matrix.

These binary matrices serve as training input for the deep neural network, representing all further processing stages beyond the DCN up to the auditory cortex and higher, language associated, cortex areas. The neural network consists of four convolutional layers and three fully connected layers, and is trained with error backpropagation on the classification of 207 different German words (Figure 1D). The resulting classification accuracy of the trained network serves as a proxy for speech recognition (Figure 1E).

In order to simulate a particular hearing loss, the output amplitudes of the cochlea model are decreased by a certain factor, independently for the different frequency channels (Figure 1F). Subsequently, these modified cochlea outputs are further processed in the LIF neurons, finally resulting in a new
binary matrix for each word for a particular hearing loss. These new matrices then serve as test data for the previously trained deep neural network, yielding a new classification accuracy. By comparing the reference test accuracy (without any hearing loss) with the new test accuracy, the effect of a particular hearing loss on speech recognition was estimated.

Optionally, Gaussian noise with zero mean and a certain standard deviation, representing somatosensory input to the DCN, was added independently to the input of each LIF neuron (Figure 1G). Here, the standard deviation corresponds to the noise intensity. As described before, again this finally results in a new binary matrix for each wave file, yet corresponding to a particular hearing loss and, in addition, also to a particular set of frequency channel specific noise intensities. Again, all these new matrices serve as test data for the deep neural network. By comparing the reference test accuracy (without any hearing loss and noise) with the new test accuracy, the effect of particular noise intensities on speech recognition with a certain hearing loss was estimated. A sketch of the complete data flow in case of certain hearing loss and additional noise is depicted in Figure 2.

Simplified Model of the Cochlea

The cochlea is simulated as 30 Butterworth bandpass filters (3rd order) with no overlapping bands. These 30 bandpass filters are a simplification of the more than 3,000 inner hair cells of the human cochlea (Dallos, 1992). In contrast to other complex cochlea models (Tan and Carney, 2003; Chambers et al., 2019), this simplification of the dynamics of the inner hair cells was chosen to derive basic principles and to increase interpretability. The center frequencies (of the bandpass filters) are between 100 Hz (minfreq.) and 10 kHz (maxfreq.) including the complete frequency range needed for speech comprehension. The center frequencies are chosen to grow exponentially \( \text{centerfreq.} = \text{minfreq.} \cdot \text{factor}^i \) with \( i \in \{0,1,\ldots,29\} \) and \( \text{factor} = \frac{\text{maxfreq.}}{\text{minfreq.}} \cdot \frac{1}{(\text{channels} - 1)} \). Thus, for higher frequencies the spacing of the center frequencies becomes larger in analogy to the tonotopy of the human cochlea (Kandel et al., 2000; Fox, 2006). The width of the bandpass filters is defined as \( \text{centerfreq.} \cdot \text{factor}^{-0.5} \), \( \text{centerfreq.} \cdot \text{factor}^{-0.5} \).

Model of the Dorsal Cochlear Nucleus

The dorsal cochlear nucleus (DCN) was modeled as 30 leaky integrate-and-fire (LIF) neurons (Burkitt, 2006), each of these neurons is connected to one frequency channel of the cochlea. Thus, no lateral inhibition was realized to focus on the core effects. The maximum spiking rate of the simulated LIF neurons is approximately 4 kHz \( (t_{\text{refrac.}} = \frac{11}{44100} \approx 11 \frac{1}{4400} \approx 0.25 \text{ ms}, \) \( t_{\text{refrac.}} \): refractory time), which is much more than the maximum spiking rate of a biological neuron (400 Hz) (Nizami, 2002). Thus, in the simulation 1 LIF neuron represent approximately 10 real neurons. The recruitment of several neurons to increase the frequency range in which phase coupling is possible is a core concept within the dorsal cochlear nucleus (Langner, 1988).
The numerical integration of the LIF neurons was performed using the “Euler” method, as this method lead to the lowest computational complexity compared to “Heun” and “Runge Kutte”—being standard integration techniques (Fathoni and Wuryandari, 2015)—without causing significant inaccuracies.

Model of Brain Stem and Cortex

The neural processing stages of the auditory pathway above the DCN including superior olive, lateral lemniscus, inferior colliculus, medial geniculate corpus in the thalamus, and auditory cortex are modeled as a deep neural network (Kandel et al., 2000). For our main simulations with the custom-made data set we used a Deep Convolutional Neural Network (LeCun et al., 2015) (for architecture and exact parameters see Table 1). For the FSDD data set we used a slightly different architecture (Table 2). Furthermore, we also used Deep LSTM networks (Hochreiter and Schmidhuber, 1997) to double-check the validity and universality of the beneficial effects of intrinsic noise (Table 3).

Data Sets

Custom-Made Data Set

Our custom-made data set created for the purpose of the present study was recorded from 12 different speakers (6 male, 6 female) in a range of 20–61 years. The data was recorded with a sampling rate of 44.1 kHz bit using Audacity. Each participant had to speak the 207 most common German words 10 times each. After recording, the data was labeled using forced alignment and cut into 1 s intervals. The data from 10 participants served as training data set, whereas the data from the two other speakers was used as test data set. All evaluations, i.e., simulated hearing loss and effect of intrinsic noise, were based on the modified test data.

Ethics Statement

All experiments were performed in accordance with relevant guidelines and regulations. Informed consent was obtained from all subjects. According to the Ethics Committee of the University Hospital Erlangen, no further ethics approval was required since non-invasive studies like this are exempted from formal ethics approvals.

Free Spoken Digit Data Set Data Set

The second used data set is an open data set consisting of spoken digits (0–9)—in analogy to the MNIST data set— in English. The data set is sampled with 8 kHz and consists of 2,000 recorded digits from four speakers (Jackson et al., 2018). Here the first five repetitions of for each speaker and each digit are used as test data, the respective remaining 45 repetitions serve as training data.

Training of Deep Neural Networks With Undisturbed Test Data

As described above the complete auditory pathway beyond the DCN, including the superior olive, lateral lemniscus, inferior colliculus, medial geniculate corpus, and the auditory cortex, is modeled as a deep neural network which is trained on the classification of 207 different German words (custom-made data set), or 10 English words corresponding to the digits from 0 to 9 (FSDD data set; Jackson et al., 2018), respectively. In both cases the compressed, i.e., down sampled, DCN output matrices served as training and test data input.

In case of our custom-made data set, the network is exclusively trained on the data of 10 out of 12 speakers, while the remaining two speakers serve as test data. Furthermore, for network training we used only those compressed spike train matrices that correspond to the undisturbed system, i.e., without hearing loss and added noise. Due to the image-like features of the compressed spike pattern matrices [similar to frequently used Mel spectrograms (Meng et al., 2019) in speech recognition, the deep neural network mainly consisted of convolutional layers. The exact architecture and all parameters are provided in Table 1]. For training on our custom-made data set, the maximum test accuracy (0.37) significantly decreases after 20 epochs of training, and thus we applied the early stopping procedure (Caruana et al., 2001) to prevent the network from overfitting. The trained networks were used for all further analyses with different modifications of the test data set, i.e., different hearing losses and different intensities of intrinsic noise.

Simulation of Hearing Loss

The hearing loss was simulated by a linear attenuation of the cochlear output at the affected frequency ranges. Thus, a hearing loss of X dB means that the outputs of the affected frequency channels are multiplied with the factor $10^{-X/20}$. Additionally, for further experiments we added a real information loss by setting an additional threshold. A threshold of $-X$ dB means that all values, where the absolute value is smaller than the threshold value $10^{-X/20}$ are set to 0 (see Figure 9A).

DISCUSSION

In this study, we demonstrated with a computational model of the auditory system that noise added to the DCN may improve speech recognition after hearing loss, by means of SR. The relative benefit of SR turned out to be largest for hearing losses between 20 and 30 dB.

Because SR works by partly restoring missing information in the input data, adding noise improves the classification accuracy of the neural network even after the training period is finished. This stands in contrast to machine learning techniques that achieve an increased robustness and generalization ability by purposefully using noisy training data from the beginning (Karpukhin et al., 2019), or by adding artificial noise during the training period (An, 1996; Zhao et al., 2019).

Furthermore, this is also the crucial difference between the SR model of auditory perception and alternative central gain models. Instead of restoring the average spontaneous neural activity after hearing loss, SR increases the information transmitted to the auditory system.

In our work, we first train the neural network for speech recognition, then simulate a hearing loss, and finally reduce this loss by adding noise. This approach is biologically plausible, as also the brain is trained on speech recognition during childhood (Dabrowska and Kubinski, 2004; Gervain, 2015), where hearing
ability is usually optimal [Indeed, hearing impairment in childhood can lead to problems in language acquisition, which cannot be fully cured in adulthood (Pimperton and Kennedy, 2012)]. In the course of a lifetime, hearing ability becomes permanently (Gates and Mills, 2005; Huang and Tang, 2010) or temporarily worse (Willott and Lu, 1982), often due to high amplitude sound exposure.

We have proposed that hearing ability can be restored by a control cycle embedded in the brain stem, along the auditory pathway, which uses internal neural noise to exploit the effect of stochastic resonance (Krauss et al., 2016). Thus, it is supposed that the neural activity in damaged frequency channels is up-regulated by internally generated noise to restore hearing within this frequency range. Indeed, simulated transient hearing loss improves auditory detection thresholds (Krauss and Tziridis, 2021).

Overshooting of this noise up-regulation is proposed to be the origin of tinnitus (Krauss et al., 2016). Our model could provide an interesting explanation for overshooting internal noise: In our simulation of high frequency hearing loss, we found that the accuracy as a function of the added noise has not only a single maximum, as expected for a resonance curve, but features a second maximum at a higher noise level (Figure 8B). If the neural control cycle would be drawn to this secondary maximum, this might explain an overshooting of the neuronal noise and the corresponding emergence of tinnitus (Krauss et al., 2016, 2017; Schilling et al., 2021d). Another potential cause of tinnitus arises from the fact that phase locking, the encoding of a signal's phase information in neural spike trains, is only possible for frequencies up to 4 kHz, the maximum spike rate of the DCN neurons (Figure 3A).

The stochastic resonance effect probably works only below this limit frequency, and thus it is not clear whether (or how) the neural control system compensates for the hearing loss in the frequency range above 4 kHz, as it has no real maximum to optimize for. Potentially, the tuning of the noise parameters in this frequency regime is done only by random trial. This model would fit to the observation that tinnitus mainly occurs in the high frequency range (Gollnast et al., 2017).

While we propose the DCN to be the place where auditory input from the cochlea is integrated with neural noise from the somatosensory system, we cannot rule out that SR rather occurs in the ventral cochlear nucleus (VCN) instead. Our LIF neurons correspond to narrow band neurons, which transform their cochlear input with minimal processing into spike trains. Neurons like e.g., bushy cells with such primary-like responses that show increased spontaneous firing rate after hearing loss are known to exist also in the VCN (Martel and Shore, 2020). In contrast, DCN neurons show strong non-linearities in sound processing through inhibitory shaping of their responses by inhibitory inter-neurons (Young and Davis, 2002; Oertel and Young, 2004). This circuitry might be the correlate of the noise-adjusting feedback-loop proposed in our model. For the sake of simplicity, we did not explicitly model this exact circuitry. Furthermore, the VCN is also innervated by trigeminal nerve fibers (Wu et al., 2015, 2016) which may be the source of the neural noise for SR. However, the DCN identification is not necessary for our model to work, and the identification of our model LIF neurons as VCN neurons would be possible as well.

We were able to show that neural noise could potentially help to increase speech comprehension in neural systems in a computational model of the auditory pathway. Even though, previous studies suggested a benefit of SR of only about 5 dB (Zeng et al., 2000; Krauss et al., 2016; Gollnast et al., 2017), an accuracy improvement of up to a factor of 2 is possible. This model provides new insights how the auditory system optimizes speech comprehension on small time scales, and why this processing was evolutionary preserved even though, tinnitus results in strong psychiatric burden: comprehension of natural speech (Schilling et al., 2021c; Garibyan et al., 2022) is essential for humans. More general, recognition of communication sounds can be assumed to be essential for all social species, in particular mammals. This may explain why behavioral and neural correlates of tinnitus are also frequently observed in rodents.

Furthermore, we could give a mechanistic explanation of the development and characteristics of tinnitus perception. These finding could have a major impact on medical treatment of phantom perceptions, but on the other hand raises new research questions in the field of engineering.

However, it has to be stated that the SR model of tinnitus development is by no means complete. While our model provides a valid explanation for acute tinnitus perceived directly after noise trauma, and also explains why a tinnitus percept could be suppressed by acoustic noise of low intensity (Schilling et al., 2021a; Tziridis et al., 2022), it does not include long-term neural circuit-level effects (Jeschke et al., 2021) due to neural plasticity. Furthermore, our model is [like the central gain (Auerbach et al., 2014) and the lateral inhibition model (Gerken, 1996)] a pure bottom-up model, which means that cortical or thalamocortical top-down modulations are not regarded. Note that, we do not discuss further bottom-up models of tinnitus development in detail, as these models make no predictions on speech perception benefit of tinnitus after hearing loss (for an in-depth comparison of the different models, see Schilling et al., 2022). In contrast to bottom-up models, top-down models could play a major role in understanding why brainstem hyperactivity passes the “gate to consciousness” (the thalamus) and results in suffering a psychic burden. Furthermore, attention effects could play a crucial role in stress related modulations of tinnitus loudness (Mazurek et al., 2015). Thus, some models describe the conscious tinnitus percept as a consequence of thalamocortical dysrhythmia. This dysrhythmia is induced by changed thalamo-cortical signal transmission, which is a result of reduced resp. increased subthalamic input to the medial geniculate body (Llínas et al., 1999; De Ridder et al., 2015; Gault et al., 2018). More recent approaches suggest that tinnitus is a prediction error and formalize their models within the Bayesian brain framework (Sedley et al., 2016; Hullfish et al., 2019; De Ridder and Vanneste, 2021). In summary, it is necessary to merge bottom-up and top-down models of tinnitus development to achieve a unified explanation of tinnitus development (Schilling et al., 2022). Our bottom-up model has not exclusively explanatory power but might also serve as source of inspiration for advanced machine learning approaches.
Furthermore, the effect of SR could be used to improve sensory systems (Krauss et al., 2017). Although noisy data is already used to make machine learning approaches more stable and less vulnerable to small distortions (e.g., Neelakantan et al., 2015; Gulcehre et al., 2016), the SR phenomenon can be used in a different way. Thus, feedback loops could be implemented in artificial intelligence systems, which are optimized on finding the ideal noise level to make a signal detectable. This approach goes well beyond already established techniques in artificial intelligence research.

Our study provides evidence that an interplay of deep learning and neuroscience helps on the one hand to raise understanding of the function of biological neural networks and cognition in general (e.g., Schilling et al., 2018, 2021b; Krauss et al., 2019a,c,d, 2021; Gerum et al., 2020; Krauss and Maier, 2020; Bönsel et al., 2021; Metzner and Krauss, 2022), an emerging science strand referred to as cognitive computational neuroscience (Kriegeskorte and Douglas, 2018). On the other hand, fundamental processing principles from nature—such as stochastic resonance—can be transferred to improve artificial neural systems, which is called neuroscience-inspired AI (Hassabis et al., 2017; Gerum et al., 2020; Gerum and Schilling, 2021; Yang et al., 2021; Maier et al., 2022).

DATA AVAILABILITY STATEMENT

The raw data supporting the conclusions of this article will be made available by the authors, without undue reservation.

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AUTHOR CONTRIBUTIONS

AS, RG, and PK performed the simulations. AS, RG, CM, and PK analyzed the data. CM and AM provided the analysis tools. PK and AM supervised the study. All authors wrote the manuscript.

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