

# An Automatic Sequential Recognition Method for Cortical Auditory Evoked Potentials

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**Abstract**—The detection of cortical auditory evoked potentials (CAEP), which are part of the electroencephalogram (EEG) in reaction to acoustic stimuli, has important applications such as determining objective audiograms. The detection is usually performed by a human operator, with support from often basic signal processing methods. This paper presents a novel mechanism for the detection of CAEPs, which is fully automatic and stops the measurement when a given confidence is reached. This proposed detector comprises of three stages. First, a feature extraction by a wavelet transform parameterizes the time domain EEG signal by only few transform coefficients. This feature vector is then classified by a neural network which yields a binary vote on every EEG segment. Finally, a sequential statistical test is performed on successive classifications; this stops the measurement if a specified decision confidence has been reached. The adjustment of the detector according to a clinical database is discussed. Thus adjusted, the proposed CAEP detection scheme is applied to a study, and compared with a human operator. The results demonstrate that this method can attain similar results, but outperforms the human expert for stimulation levels close to the hearing threshold.

**Index Terms**—Automatic detector, cortical auditory evoked potentials, neural networks, sequential statistical test, wavelet analysis.

## I. INTRODUCTION

CORTICAL auditory evoked potentials (CAEPs) can be observed as part of the electroencephalogram (EEG) and reflect the neural processing of acoustic stimuli in the cortex. In their clinical application, CAEPs have the advantage of enabling frequency specific testing of the hearing ability by appropriate selection of the stimulus [1]. The recognized presence or absence of a CAEP in response to an acoustic stimulus of specified frequency and level can be used to determine an objective audiogram. Objectivity is achieved in the sense that the active cooperation of an experimentee or patient is replaced by the interpretation of the EEG by an operator [1]. Objective hearing assessment is important to, for example, establish reliable hearing

thresholds for persons with low compliance, e.g., small children [2]. The work presented here is concerned with an automatic detector to help and assist in identifying CAEPs.

When an acoustic stimulus reaches the inner ear it elicits neural activity along the neural auditory pathway from the spiral ganglion via nerve fibers over the brainstem up to the auditory cortex, where it results in auditory perception approximately 100 ms after the onset of the stimulus [3]. Triggered by processing in the auditory cortex [4], CAEPs, therefore, give evidence of correct auditory perception. According to [5] and references therein, for most human adults the signal properties of the CAEP are quite similar when the same measurement paradigm is used and the central auditory processing is normal. The CAEP consists of waveforms with specific latencies with respect to the stimulus onset. The literature names their minima and maxima due to polarity and sequence  $P1$ ,  $N1$ ,  $P2$ , and  $N2$ , whereby  $N1$  and  $P2$  are the most stable waves of a CAEP with peak latencies of about 100 ms and 175 ms, respectively [5]. It is generally the recognition of this  $N1$ - $P2$ -complex which is used to establish one of two hypotheses  $H_0$ : no CAEP within EEG; and  $H_1$ : CAEP within EEG [6].

Variations of amplitudes and latencies of the CAEP waves occur and depend—besides age, sex, ear of stimulation, hearing ability, and vigilance of the experimentee—on the frequency, level and repetition rate of the stimuli [5]. While many of these factors only contribute little or can be controlled during the measurement, the individual sensation level is in general unknown and has an observable influence: As the stimulus level approaches the hearing threshold, the CAEP waves appear weaker and more delayed due to prolonged auditory processing for quieter sounds [3]. An example is given in Fig. 1, where averaged EEGs recorded synchronously to stimuli of different intensities are shown.

Besides this variability, detection of CAEPs is impeded by background EEG due to visual, motoric, and sensoric activities of the experimentee, resulting in a very low signal-to-noise ratio (SNR) of less than  $-10$  dB. As CAEP and background EEG cover similar frequency bands, it is impossible to detect a CAEP from EEG measurements with only a single stimulus. Rather, the detection of CAEPs generally has to be based on repeated stimuli and the acquisition of typically about 30 to 80 “sweeps” which, here, refer to EEG segments recorded under identical conditions synchronously to the given stimulus. Many methods for support in this decision have been proposed, and range from simple averaging to more advanced signal processing.

Simple averaging of successive sweeps is a common, but often sub-optimal CAEP reconstruction method, since stationarity of the background noise, invariance of the CAEP, and

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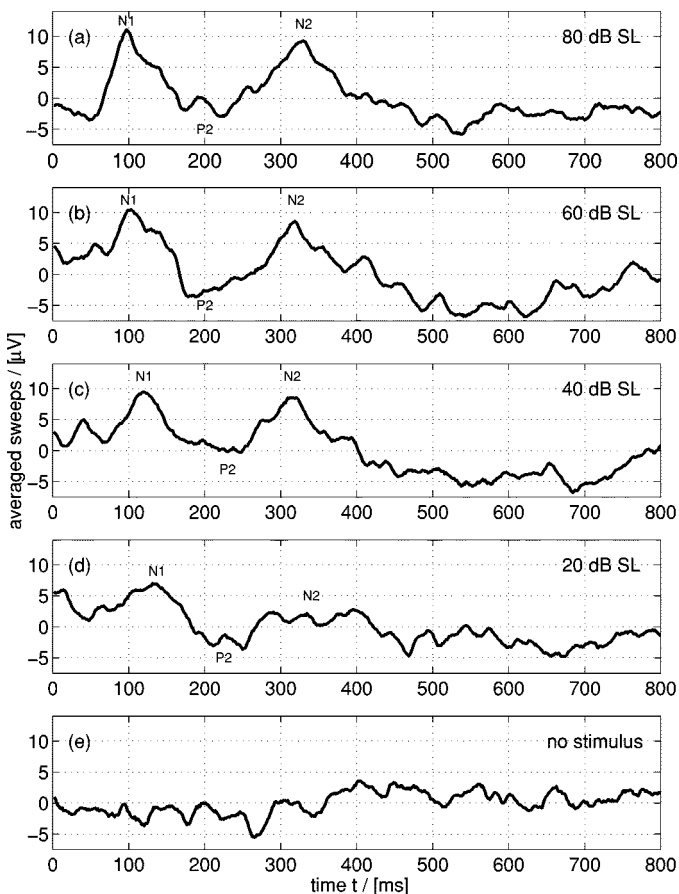


Fig. 1. EEG averages over 100 sweeps obtained from a single individual: (a)–(d) variable stimulation intensities of 80, 60, 40, and 20 dB above hearing threshold (SL) applied at  $t = 0$  containing CAEP; (e) no stimulation ( $-\infty$  dB) and, hence, no CAEP.

the absence of correlation between CAEP and noise have to be assumed. Due to varying vigilance and habituation effects during the measurement, this is only approximately satisfied [7]. Methods have been proposed to take variations of the background EEG into account by estimating the noise power in every sweep [8]–[11], and weighting their contribution to the average accordingly. Thus, noisier sweeps and artifacts are deemphasized. Other approaches try to enhance the SNR by considering the variations of the central auditory processing during the measurement [12]–[14]. These techniques assume that the latencies of the CAEP vary over time, leading to an approximate affine translation of the signals. The actual translations are estimated by cross correlation of every single sweep with predefined patterns. SNR enhancement is then achieved by averaging the latency-corrected sweeps. Clearly limited by the small sample size for variance-type estimations, such methods can only be moderately successful.

A number of parametric approaches aim to replace the subjective interpretation of the averaged CAEP by an automatic classification, which is generally based on the statistics of some sweep parameters. Literature offers a wide range of parameters being used, like CAEP relevant samples of the digitized signals, selected Fourier coefficients, energy of averaged sweeps, or correlation coefficients with “normal” CAEP [15]. Coarse SNR estimates of the CAEP/EEG based on the comparison of a stan-

dard average with a “ $\pm$  average,” where every second sweep is included with reversed sign, were also used to classify CAEPs [16], [17], [15]. Wavelet coefficients have been used to parameterize evoked potentials successfully [18], and have been shown to perform at least comparable to the above parametric methods for CAEPs [19]. However, such parametric approaches require to *a priori* set a threshold for classification and cannot provide stopping criteria for the measurement based on a specified error rate, which usually is regarded as major drawback in clinical use.

In the approach proposed here, the good parameterization property of the wavelet transform is exploited to build a detector with stopping criterion and specified error rate. This detector is outlined schematically in Fig. 2. First, a selected set of discrete wavelet transform (DWT) coefficients, a so called “feature vector” (FV), is calculated for each measured sweep. The dimension of the FV can be much smaller than the original time-domain signal without loss of too much information on a CAEP [18], [19]. The FVs of successive sweeps are each classified by a neural network (NN), which maps the FV onto a single binary  $\{0, 1\}$  output. A similar approach is implemented by wavelet networks [20], which combine both a wavelet transform for parameterization and an NN in a hybrid classifier. As the binary classification of the NN bears a high type-I-error (incorrect acceptance of  $H_1$ ), a sequential statistical test forms the last stage of the detection procedure, judging the number of positive and negative sweep classifications produced by the NN. This test will provide a decision based on a predefined type-I-error rate, and stops the measurement when this confidence limit is reached.

In Section II, the proposed detector is introduced in detail. Section III reports on the EEG equipment at hand and the resulting data properties, which are used to adjust the different stages of the detector. This includes the selection of the FV, the dimensioning and training of the NN, and the adjustment of the sequential statistical test. The application of our method to a study using clinical data is reported in Section IV, compared to the results achieved by a human operator. Finally, we draw conclusions in Section V.

## II. AUTOMATIC DETECTOR METHODS

This section describes in detail the principles of the proposed detection scheme and its components. The parameterization by DWT coefficients is addressed in Section II-A. Section II-B introduces a NN classification of the features, thus, extracted by the DWT. Finally, a sequential statistical test to assess the classification results of successive sweeps is introduced in Section II-C. The adjustment of the detector components with respect to EEG/CAEP data will be performed in the separate Section III.

### A. CAEP Feature Extraction by Wavelet Transform

To parameterize a CAEP and condense the information on whether an EEG contains or does not contain a CAEP within only few coefficients, a wavelet transform is invoked. Similar to the Fourier transform, the wavelet transform performs a least-squares fit of an analysis function to the time domain data. However, rather than a sinusoid extending over an infinite interval, here wavelet functions with a limited support are employed as

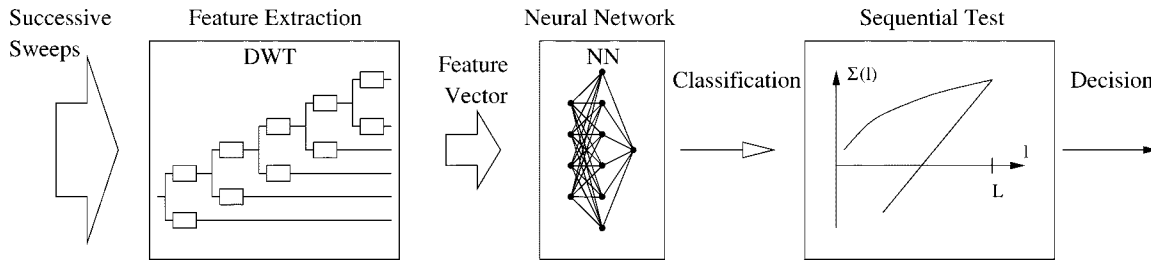


Fig. 2. Block diagram of the CAEP detection scheme.

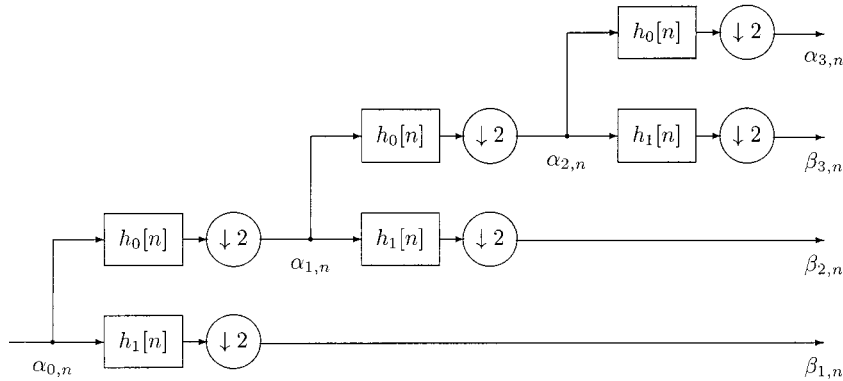


Fig. 3. Octave filter bank to compute an MRA of depth  $D = 3$ ; deeper decompositions are achieved by further splitting  $\alpha_{3,n}$ .

basis functions, which is advantageous when analyzing transient signals [21]. For the DWT considered here, a set of orthogonal basis functions is obtained by scaling and translation of a mother wavelet.

An efficient calculation of the DWT coefficients in the case of discrete-time data can be achieved with a multiresolution algorithm (MRA, [22]). For a dyadic DWT, this MRA is performed by filtering the signal to be analyzed with an octave filter bank as shown in Fig. 3. The high-pass filter  $h_1[n]$  forms a quadrature mirror filter (QMF) pair [23] with the low-pass  $h_0[n]$  of the filter bank. The input sequence to the octave filter bank,  $\alpha_{0,n}$ , is the signal  $x[n]$  to be analyzed. Through successive low- and high-pass filtering of the samples in the lower frequency band,  $\alpha_{m,n}$ , and decimation of the resulting signals by a factor of two (denoted as  $\downarrow 2$ ), subband samples  $\beta_{m,n}$  are obtained, which, except of the lowest frequency band, represent the DWT coefficients. The coefficients  $\alpha_{m,n}$  are intermediate values and correspond to a dual basis function of the wavelet, a so called scaling function [22]. The filters  $h_0[n]$  and  $h_1[n]$  are sampled version of the underlying scaling function and wavelet and, therefore, determine which DWT—amongst a large variety of possible wavelet functions (see, e.g., [22], [21], and [24])—is being implemented.

Since the signal to be analyzed,  $x[n]$ , is only defined on a finite interval  $n \in [0; N - 1]$ , suitable signal extensions [25] or boundary filters [26] have to be selected. As both yield identical results [26], for ease of implementation the first possibility will be pursued. Zero padding is not viable for decomposition with a deep octave bank, as information will be blurred over large intervals by the transient behavior. Periodization of the original data and all intermediate filtering results  $\alpha_{m,n}$  however permits to filter in steady-state and, thus, retain all required information within a preserved period of the signal. With the possibilities of odd or even periodization [25], the MRA yields  $N$  DWT coeffi-

cients that completely describe the original time series  $x[n]$  comprising  $N$  samples. However, restrictions to symmetric wavelets apply, if an odd periodic extension is selected [22].

Based on EEG data, Section III will determine, which wavelet function and periodic extension will be used for the proposed automatic detector and which coefficients can be isolated as significant. The task will be to identify a subset of  $M \ll N$  coefficients collected in a FV,  $y_{FV}$ , that can describe the CAEP sufficiently well. Section III will also show why these coefficients on their own can only yield a rather unsatisfying indication of a CAEP. This will motivate the requirement of the NN classification and sequential test performed on the selected significant DWT coefficients in the FV described in Section sII-B and Section sII-C to ensure a reliable and systematic CAEP detection.

### B. NN Classification

The aim of this section is to introduce a classification method for single sweeps distinguishing between “containing a CAEP” or “not containing a CAEP” based on the set of  $M$  DWT coefficients in the FV. Due to interindividual variations of CAEP-latencies and -amplitudes these FVs do not exhibit a single specific pattern. Instead, many patterns must be regarded as being specific for CAEPs. So far, there exists no theoretical treatment of this discrimination problem; hence, heuristic classifiers have to be used. Artificial NNs provide the possibility of learning by groups of sample FVs and require only little computational effort.

A multilayer perceptron was used for classification of the FV calculated for each sweep. This type of NN comprises an input layer with  $M$  nodes corresponding to the dimension of the FV, one hidden layer with  $K$  neurons and an output layer with a single neuron ( $M - K - 1$  network). Since the NN is a nonlinear processor, a bias removal and normalization with respect to the

$l_\infty$ -norm are applied to the FV  $\mathbf{y}_{FV}$  prior to feeding it into the NN. This operation excludes interindividual amplitude differences that might be incurred due to slightly different measurement impedances and anatomical differences. The resulting modified FVs  $\tilde{\mathbf{y}}_{FV}$  now lie on the surface of an  $M$ -dimensional hypercube.

While the number of nodes  $M$  in the input layer is determined by the dimension of the FV, the number of neurons in the hidden layer,  $K$ , has to be selected appropriately. If  $K$  is chosen too small and not enough hidden neurons are used, during training the NN is unable to classify all FV of the learning set of data. On the other side, a too large number of hidden neurons  $K$  will cause the NN to be too specific for the learning data set, and prevent the NN to be able to interpolate (or “generalize”) when slightly different CAEP data is presented for classification.

The classification process for a FV  $\tilde{\mathbf{y}}_{FV}$ , where  $l$  is the sweep index, can be summarized as

$$\gamma(l) = s(\mathbf{b}_2 + \mathbf{W}_2 \cdot \tanh(\mathbf{b}_1 + \mathbf{W}_1 \cdot \gamma \tilde{\mathbf{y}}_{FV}(l))) \quad (1)$$

where  $\tanh$  is a vector valued hyperbolic tangent and  $s(x)$  a Heavyside-function [ $s(x) = 1$  for  $x \geq 0$ ,  $s(x) = 0$  else] to produce a binary classification  $\gamma(l) \in \{0; 1\}$ . During the training phase of the NN the parameters in weight matrices  $\mathbf{W}_1$ ,  $\mathbf{W}_2$ , and bias vectors  $\mathbf{b}_1$ ,  $\mathbf{b}_2$  are adjusted such that the classification error for certain training FVs will be minimized. This can be done by the backpropagation algorithm<sup>1</sup> [27], [28]. The description of the FV set used for the NN training to appropriately select  $K$  and adjust the weight and bias parameters in (1) will be described in detail in Section III-C.

It is obvious that this classification based on single sweeps bears large classification errors. This is unavoidable as the time signal in the spontaneous EEG is known to often exhibit patterns similar to a CAEP. In Section III-C, we will establish the exact probability  $p_{NN}$  for false detection of a CAEP by the appropriately dimensioned and trained NN. This error rate  $p_{NN}$  can be interpreted as the fractional area on the surface of the  $M$ -dimensional hypercube occupied by those FVs  $\tilde{\mathbf{y}}_r(l)$  which will be classified as a CAEP.

The relatively high false positive rate that will be encountered here is due to the low SNR of the CAEP within the spontaneous EEG. Enhancement of this SNR could be achieved by sub-averaging over two or more sweeps with subsequent NN-classification of the FV of the averaged data. This, however, has drawbacks. Firstly, this procedure prolongs measurement time proportionally. Secondly, although the false negative error is reduced, the false positive error classification  $p_{NN}$  remains unaffected. A better approach can be achieved by sequential statistical analysis of the NN classification results. The statistical method to perform this final stage of the detector is outlined next.

### C. Sequential Statistical Analysis

To assess the classification results yielded by the NN in Section II-B, usually the statistics of a fixed number  $L$  of sweeps is tested. Based on this standard method introduced in Section II-C1, in Section II-C2 we derive a sequential test based

<sup>1</sup>To apply the algorithm, for differentiability during training the output nonlinearity  $s(x)$  is replaced by a hyperbolic tangent; subsequently the desired output of the NN during training was in [1;1].

on [29], [30], which only defines an upper limit of sweeps,  $L$ , but can be stopped earlier if the statistics agree.

1) *Statistical Test with Fixed Sample Size*: Consider a random series of EEG sweeps measured without any acoustic stimulation and, therefore, without any CAEP. The classification results of the subsequent FVs by the NN gives a sequence of zeros and ones, whereby the probability to misclassify such FVs as to “containing a CAEP” ( $\gamma(l) = 1$ ) is denoted by  $p = p_{NN}$ . After  $L$  recorded sweeps the sum  $\Sigma(L)$  of all classifications

$$\Sigma(L) = \sum_{l=1}^L \gamma(l), \quad \gamma(l) \in \{0, 1\} \quad (2)$$

lies between zero and  $L$ . Since the single classification results are independent, the sum  $\Sigma(L)$  is binomially distributed. Thus, the problem is to test the null hypothesis  $H_0: p = p_{NN}$  (no CAEP is inside the EEG) against the alternative  $H_1: p > p_{NN}$  (a CAEP is present). The binomial distribution can be approximated for sufficiently large  $L$  by a Gaussian distribution with parameters

$$\mu(L) = L \cdot p_{NN}, \quad \sigma(L) = \sqrt{L \cdot (1 - p_{NN}) \cdot p_{NN}}. \quad (3)$$

Once the error rate  $p_{NN}$  for the NN classification in Section II-B is established from test data, the probability for  $\Sigma(L) > \mu(L) + z_\alpha \cdot \sigma(L)$  after  $L$  classifications can be set to take on a fixed value  $\alpha$  by selecting the quantity  $z_\alpha$  appropriately. Using the correct values for  $\mu$  and  $\sigma$  instead of the Gaussian approximations in (3),  $z_\alpha$  will become slightly smaller. Hence, the approximation is conservative in the sense that the true value for  $\alpha$  is slightly smaller when based on the assumption of Gaussianity. With these considerations, a statistical test with a predetermined error of false positive classifications (type-I-error)  $\alpha$  can be created.

Using this statistical test after a fixed sample size of  $L$  sweep classifications,  $H_0$  can be accepted or rejected based on whether the distribution is inside or outside the expected interval with a prespecified confidence. However, it is important to note that this test only works with a fixed number of sweep classifications.

2) *Sequential Statistical Test*: Sequential testing means that although the sample size has an upper limit,  $L$ , the sum  $\Sigma(l)$  of NN classifications

$$\Sigma(l) = \sum_{\lambda=1}^l \gamma(\lambda), \quad l \in [1; L] \quad (4)$$

where  $l$  is the sweep index, is checked during the entire measurement. The measurement will be stopped immediately if certain statistical limits are exceeded, resulting in an acceptance of  $H_1$ .

Due to the complex distribution of  $\Sigma(l)$ , the statistical limits cannot be directly calculated. Therefore, they were determined using Monte-Carlo simulations based on test data, which are described in detail in Section III.

*Example*: An example expressing the main idea of the sequential test is shown in Fig. 4. There, the abscissa represents the number of single sweep measurements,  $l$ , the ordinate the sum  $\Sigma(l)$  of binary NN classifications. The upper limit in this example is  $L = 75$ . If after  $l$  sweeps  $\Sigma(l)$  exceeds the upper boundary, a CAEP is detected and the measurement will be

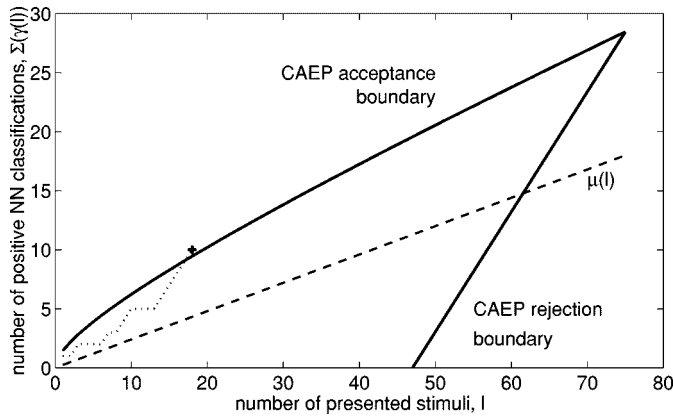


Fig. 4. Sequential test plan with acceptance and rejection boundaries for assessment of NN classification ( $L = 75$  and  $p_{NN} = 0.24$ ); dashed line: mean value for the sum of positive NN classification for random sweeps; dotted line: example of sweep with CAEP detected after  $l = 18$  NN classifications.

stopped. If the lower boundary is reached after  $l$  sweep recordings, it becomes impossible that the upper boundary is met in the remaining  $L - l$  sweeps, a CAEP is rejected and the measurement can be stopped. The dashed line in Fig. 4 shows the mean value  $\mu(l) = l \cdot p_{NN}$  for positive NN-classifications in the absence of a CAEP (type-I-error). An example for  $\Sigma(l)$  is given with the dotted line which, here, passes the acceptance boundary ( $\Sigma(18) = 10$ ) after  $l = 18$  sweep classifications, and a CAEP is detected. It can be easily calculated that a minimum number of  $l = 3$  sweep classifications is required before a CAEP can be detected under the conditions of this example.

The above sequential test procedure can be regarded as a semi-adaptive decision algorithm [29]. If the SNR of the CAEP inside the ongoing EEG is high enough, the algorithm will considerably shorten the time which is necessary to reach a decision. However, if the SNR is extremely low, a larger number of measurements will be required. In case of the example given above, the test will need an average number of 62 sweeps until a measurement will be rejected as non-CAEP, as indicated by the intersection of the mean sum  $\mu(l)$  with the rejection boundary in Fig. 4.

### III. ADJUSTMENT OF DETECTOR USING REAL DATA

Now, step by step, the general automatic detector outlined in Section II will be adjusted with respect to our data acquisition equipment and procedure, which is discussed in Section III-A. This has consequences for the tailoring of the DWT and the selection of transform coefficients parameterizing CAEPs in Section III-B. The subsequent dimensioning and training of the NN classifier is described in Section III-C. The false positive classification rate of the NN together with a selected significance finally sets the parameters for tuning the sequential test in Section III-D.

#### A. Equipment and CAEP Data

All EEG measurements employed in this work were obtained under clinically proven conditions in an electrically insulated and acoustically anechoic chamber [31], where patients and experimentees are seated in a comfortable armchair in order to

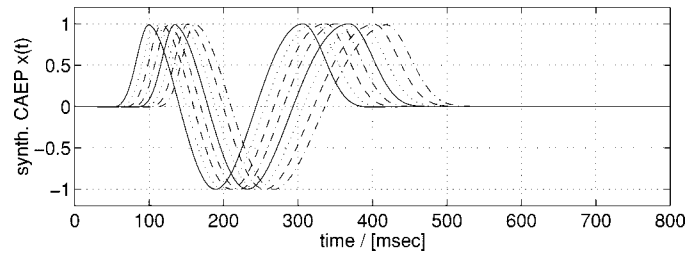


Fig. 5. Examples from the ensemble of synthetic CAEPs for additional NN training.

minimize motoric artifacts inside the EEG. The vigilance of persons under test is ensured by monitoring with a video system. The stimulation system (ASTI20, ZLE Corp., Munich) used here produces predefined signals, which are presented via a headphone (Beyerdynamics DT48) to the test persons and patients. The EEG is measured as a voltage from the ipsilateral mastoid (M1, M2) and a contralateral forehead position (FP1, FP2) of the stimulated ear against a ground electrode positioned on the center of the forehead (FZ). After differential EEG amplification by approx. 80 dB, the analog EEG signal is prefiltered by Bessel filters of 4th order with a passband region between 1.6 Hz and 20 Hz. Finally, the stimulation system triggers a 12-bit analog-to-digital converter sampling at 640 Hz for synchronous digital recording, yielding  $512 = 2^9$  samples for 800 ms segments of EEG represented within each sweep.

Sine-bursts of 300-ms duration at frequencies of 500 Hz, 1 kHz, 2 kHz, and 4 kHz were used as stimuli, with intensities set to 20-, 40-, 60-, and 80-dB sensation level (SL) for individual experimentees. For measurements without CAEP, the stimulus could also be set to zero. The stimuli were separated by randomized inter-stimulus intervals of about 1–2 s to avoid artifacts through periodic repetition. For each stimulus situation a set of at least 75–100 sweeps was recorded. Hence, a complete measurement for one stimulus situation—defined by the stimulus frequency, stimulus level and the side of the stimulated ear—extends over a duration of 2–3 min.

A set of 600 real CAEP measurements recorded with the above equipment was available for the adjustment of the detector. These measurements had been performed in the clinical routine on adult patients with a known psychoacoustic hearing threshold. The presence of CAEPs had been clearly identified by a human expert. The selection of the 600 measurements was done such that stimulation levels between 20- and 80-dB SL were uniformly represented.

Besides the real measurements, 100 synthetic CAEP time series were additionally produced for the later training of the NN. These simulated CAEPs consisted of a combination of subsequent sine and cosine half-waves with different positions of the maxima and different time scales [32], with examples given in Fig. 5.

#### B. Feature Vector

To produce a suitable FV, a number of wavelets and signal periodizations for the MRA have been tested on the data set to evaluate parameterization. The test was performed such that transform coefficients were monitored across each set of sweeps, and the relative frequency (distribution) of each coefficient was

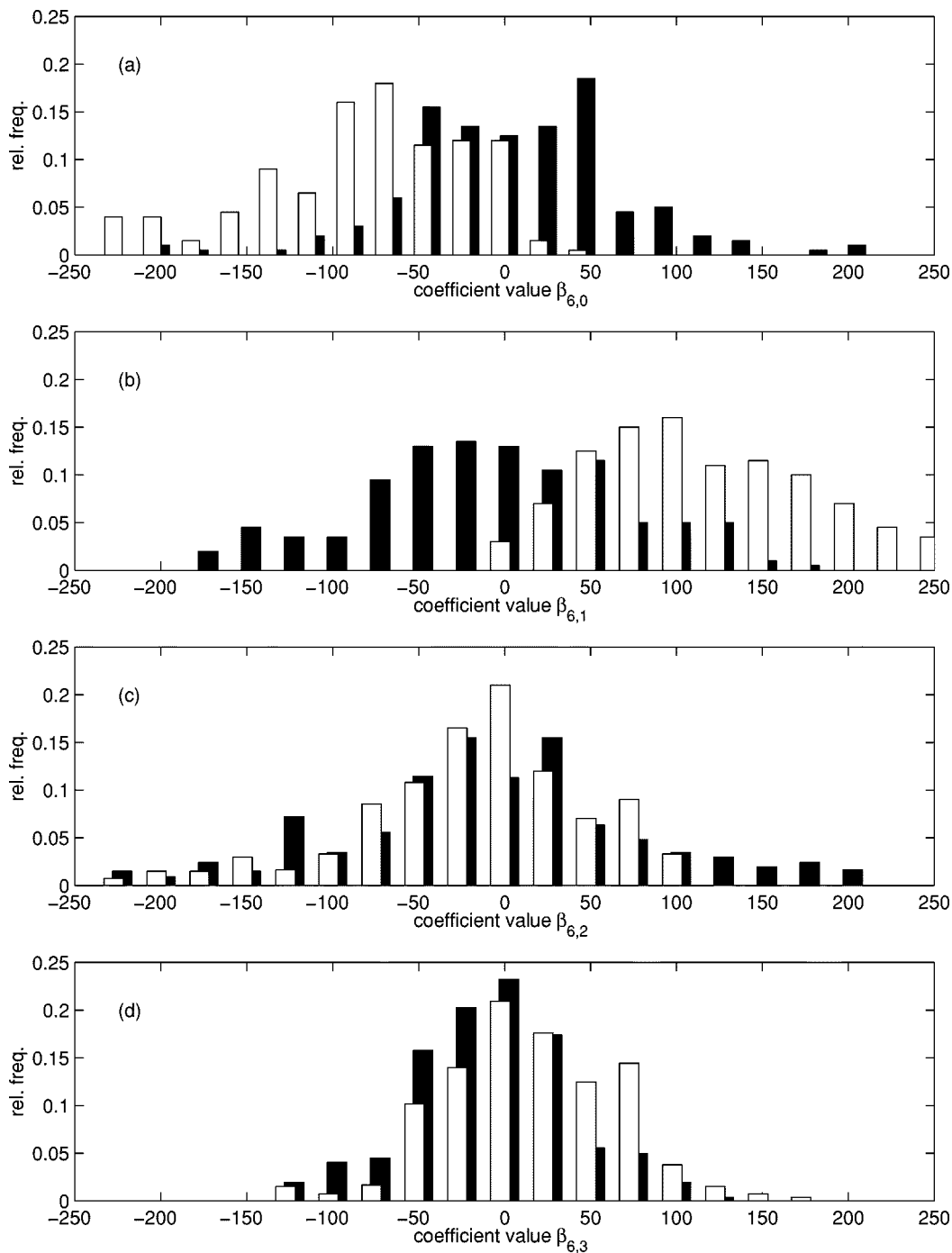


Fig. 6. (a)–(d) Frequency distributions of coefficients  $\beta_{6,0}$ ,  $\beta_{6,1}$ ,  $\beta_{6,2}$ , and  $\beta_{6,3}$ , respectively, averaged over eight successive sweeps; white histograms refer to stimulus-synchronized sweeps (1 kHz, 80-dB SL), while black bars belong to coefficient values of data without given stimulus.

compared to the distribution of an EEG measurement with no CAEP. The comparison was realized by taking normally averaged and  $\pm$  averaged sweeps, whereby the average was taken over a small sample size of eight successive sweeps. A wavelet with good parameterizing properties would then have at least one coefficient, for which a clear difference in the distributions for  $+\pm$  averaged sweeps could be measured [19].

From standard Daubechies wavelets, near-symmetric wavelets [21], and Mallat's wavelet [22], such testing gave best results for the Mallat wavelet with odd periodic extension of the data. This preference for odd extension is likely to be due to the

important components of a CAEP being close to the beginning of the sweep interval and, therefore, sensitive to discontinuities and wrap-around experienced with even periodic extensions [33]. Testing for other symmetric wavelets apart from Daubechies-2 (Haar) and Mallat wavelets was extended to a class of symmetric biorthogonal wavelets [24] which, however, did not produced better results than Mallat [34], [19].

*Example:* An example of the distributions measured for four different Mallat-DWT coefficients ( $\beta_{6,0}$ ,  $\beta_{6,1}$ ,  $\beta_{6,2}$ , and  $\beta_{6,3}$ ) for a measurement with a stimulus level of 80 dB above hearing threshold is shown in Fig. 6. Obviously, some coefficients differ

in their distributions depending on the presence of a stimulus (here,  $\beta_{6,0}$  and  $\beta_{6,1}$ ) while for others (here,  $\beta_{6,2}$  and  $\beta_{6,3}$ ) no difference can be established. In this case, the first two coefficients would be marked as “significant” for a CAEP.

With a measure how well the two distributions separate [19], not only was the Mallat wavelet determined amongst the library of wavelets tested over the data set, but the marking of “significant” coefficients would be mainly restricted to a subset of seven coefficients  $\beta_{m,n}$

$$\mathbf{y}_{FV} = [\beta_{8,0}, \beta_{7,0}, \beta_{7,1}, \beta_{6,0} \cdots \beta_{6,3}]^T. \quad (5)$$

This FV  $\mathbf{y}_{FV}$  was optimized over all sweeps in the data set, and represents approximately the first 400 ms and the frequency interval  $f \in [1.25 \text{ Hz}; 10 \text{ Hz}]$  of a sweep. As this covers both the latencies and the frequency range of 3–6 Hz generally associated with CAEPs [15], [35], the selection in (5) can be considered robust in capturing the required information. However, not all coefficients in (5) have to be indicative for a CAEP in an individual measurement, as the above example demonstrated.

An initial detector [34] was based on measuring the separability of distributions of  $+\pm$  averages over six to ten sweeps. This “separability” was based on the overlap and relative distance of mean values of the distributions. If the separability for any coefficient of the set was above a predetermine threshold, a CAEP would be assumed. Over the data set, this gives a false positive error rate of approx. 9.8%. This method, however, requires an optimum *a priori* threshold selection, has no known confidence, and needs a fixed set of measurements [19]. Here, we use this experience as a motivation for the wavelet and coefficient selection for the FV, but want to use the systematic NN classification and sequential testing, which will be adjusted next based on the properties of the FV derived over the data set in Section III-A.

### C. Dimension and Training of the NN

The FV of seven DWT-coefficients isolated in Section III-B encompasses the time-frequency range important for CAEPs and, therefore, can form the basis for further classification. Since the amplitude of the bioelectric brain signals largely depends on uncontrollable individual parameters of the test person’s vigilance and electrode impedances, no reliable information regarding CAEPs can be derived from the modulus of specific coefficients. Therefore, each FV was modified as described in Section II-B. The training of the NN was performed with FVs  $\tilde{\mathbf{y}}_{FV}$  calculated from the 600 CAEP measurements and from 100 simulated CAEP time series, as outlined in Section III-A. Besides these FVs containing a CAEP, the learning group included 700 random FVs representing measurements without CAEP.

Different numbers of hidden neurons were used for the training in order to separate the CAEP from the non-CAEP in order to achieve best performance in the sense outline in Section II-B. With  $K < 8$  hidden neurons the classification for the training data was poorer in the sense that not all of the

FVs containing a CAEP were correctly classified. With  $K = 8$  hidden neurons 100% of the CAEP and about 76% of the FV from the non-CAEP set were correctly classified. A further increase in the number of hidden neurons,  $K > 8$ , gave better results for the part of the learning set not containing a CAEP, but would decrease generalization due to the limited amount of sample FVs. The excitation of the NN with a restricted set of training data leads to adaptation of the NN parameters to this special data and is, therefore, only optimal for the training case. Hence, a number of  $K = 8$  hidden neurons was selected as optimal choice for our classification problem.

In order to evaluate the precise false positive rate of the NN, Monte-Carlo simulations were performed using  $10^4$  random input vectors. The positive classification rate for these FV was determined to be  $p_{NN} = 0.24$ , and proved to be very stable over repeated trials of this experiment.

### D. Sequential Test Setup

Once the NN is trained, the false positive rate,  $p_{NN} = 0.24$ , is fixed, as determined in Section III-C. The sequential test has to check whether the actual positive rate of the NN,  $p$ , is significantly higher than the rate  $p_{NN}$ . Two parameters of the sequential test proposed in Section II-C have to be preset. As maximum number of NN-classification,  $L = 75$  appears a reasonable number in the light of the usual 30–80 sweeps required for averaging CAEP measurements. The false positive rate (significance  $\alpha$ ) of the complete sequential test was selected to be  $\alpha = 0.05$ .

With  $\alpha = 0.05$  and the known false positive rate  $p_{NN} = 0.24$  of NN classifications,  $z_{0,05}$  can be adjusted. A rough and very conservative estimate for  $z_{0,05}$  can be derived using the Bonferroni-correction [36], resulting in a approximation  $z_{0,05} = 3.02$ . A tighter approximation for  $z_{0,05}$  can be found by Monte Carlo simulation.

For the Monte Carlo simulation, successive NN-classifications of an EEG measurement with no stimulation and  $p_{NN} = 0.24$  were simulated by drawing from a random variable uniformly distributed on the interval  $[-0.76; 0.24]$  and applying the Heavyside function. A resulting value of “0” represents “no CAEP detected,” while “1” represent “CAEP detected.” With a total number of 500 000 random series, each containing  $L = 75$  dichotome classification results  $\{0; 1\}$ , sequential tests were simulated with an *a priori* probability for positive classifications of  $p_{NN} = 0.24$ . By running the factor  $z_{0,05}$  from zero to five in steps of 0.001,  $z_{0,05}$  was determined such that the sum  $\Sigma(l)$  of NN classifications exceeds the limit at least at one time  $l \in [1; L]$  in 5% of all series

$$\Sigma(l) \geq \mu(l) + z_{0,05} \cdot \sigma(l) \quad (6)$$

where  $\mu(l)$  and  $\sigma(l)$  are defined as in (3) with  $l$  substituted for  $L$ . In order to satisfy a 5% type-I-error probability, the numerical simulation resulted in the choice  $z_{0,05} = 2.83$ . Repeated Monte Carlo trials resulted in variations of  $z_{0,05}$  of less than 0.01. Hence, in the proposed CAEP detector, the parameter  $z_{0,05} = 2.83$  will be used in the sequential test.

#### IV. CLINICAL EVALUATION AND DISCUSSION

The proposed sequential detector adjusted in Section III was evaluated in a clinical study, which is discussed in the following. Section IV-A introduces the study group and the measurements taken. The automatic detector is compared to a human expert in CAEP classification in an experiment, for which the conditions are set out in Section IV-B. Section IV-C presents and discusses the results.

##### A. Subjects

Eleven normal hearing volunteers (five male, six female) took part in the study. The experimentees were students at the University of Erlangen-Nürnberg with an average age of  $24.5 \pm 2.3$  years. The subjects had no indication of hearing deficiencies or neurological diseases and did not take any drugs the day prior to investigation. Their normal hearing status was established by pure tone audiometry, where the hearing loss at all frequencies between 125 Hz and 10 kHz was less than 10 dB. Additionally, psycho-acoustical thresholds for the stimulus levels used to evoke CAEPs were determined with the same apparatus. The stimulus levels mentioned in the following represent the levels with respect to the individual hearing thresholds, indicated by decibel sensation level (dB SL).

A total of 32 EEG measurements were obtained from each experimentee, to account for all combinations of four different stimulus level (20-, 40-, 60-, and 80-dB SL), four different frequencies (0.5, 1, 2, and 4 kHz), and the testing of both ears. Each measurement consisted of 75 EEG sweeps. In addition to these regular CAEP measurements, for each subject a number of 32 sets of EEG recordings were taken without any auditory stimulation but under otherwise identical conditions. For this case, again a minimum number of sweeps of 75 was recorded. Finally, all measurements were randomly ordered to exclude inference from previous decisions on the current measurement under test.

##### B. Test Setup and Comparison with a Human Operator

For the study, the available measurements from the subject group introduced in Section IV-A totaled 352 data sets from EEG measurements with CAEP (stimulation above hearing threshold) and further 352 data sets without CAEP due to no given stimulus. Since the complete sets of single sweep responses were stored during the EEG recordings, the whole evaluation procedure can be performed off line. Thus, the above measurements were inspected 1) with the automatic method described in this paper and 2) by an experienced human operator. From the automatic CAEP recognition both the number of necessary stimulations for sequential detection and the result of the procedure (CAEP/no CAEP) were stored.

The human operator to evaluate the data was a member of scientific staff at the University hospital in Erlangen with 15 years of experience to classify CAEPs for objective hearing assessment in the clinical routine. This human operator was placed in a fully realistic situation for EEG recordings: for each stimulus condition (stimulated ear, stimulation level and frequency) the averages of subsequent sweeps in a measurement were calculated and displayed on a PC monitor. The timing of the display was chosen similar to the interstimulus intervals during real

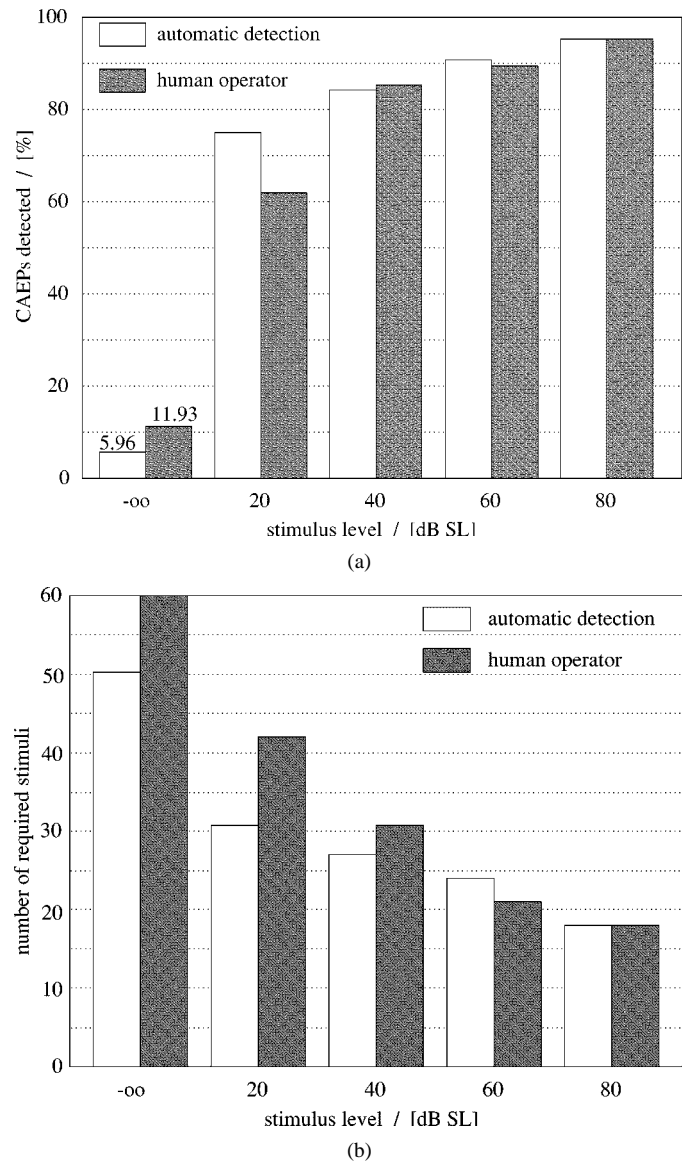


Fig. 7. (a) Decision rates for the automatic detector and those of an experienced human operator; (b) mean required number of sweeps in a measurement until a decision has been reached; a level of  $-\infty$ -dB SL refers to no stimulus given.

measurements. However, the human operator was blind to the actual stimulus level of the presently displayed measurement and he had no information about the AEP at higher/lower level for the same subject. As soon as the operator was able to reach a decision, he could stop the measurement simulation, and both his decision and the required number of stimuli were recorded.

##### C. Results and Discussion

Fig. 7(a) shows the decision results for the two methods—automatic detection and experienced human operator—on the data sets of the subject group described in Section IV-A. The average number of sweeps required by both the detector and the operator to reach a decision is given in Fig. 7(b). The results have been displayed separately for different stimulus levels, with either no given stimulus, or stimuli ranging in intensity between 20 and 80-dB SL in steps of 20 dB.



For the automatic detector, the type-I-error rate had been preset to 5% ( $\alpha = 0.05$ ) as described in Section III-D. This closely agrees with the reached average false positive classification (i.e., an CAEP is incorrectly assumed) of 5.9% for measurements with no given stimulus. It is clear from Fig. 7(a) that the human operator has a considerably higher type-I-error rate, even though he had been informed that half the measurements in the database contained no CAEP. This difference was evaluated as significant with the one-sided McNemar-Test [37] on a significance of 5%.

Looking at different stimulus levels, Fig. 7(a) shows a clear increase in the detection rate of CAEPs with the stimulation intensity for both the automatic sequential detector and the human operator, while the corresponding required measurement times in Fig. 7(b) decrease. This is due to the fact that stronger stimuli yield clearer pronounced CAEPs with larger amplitude and, thus, higher SNR, as shown in Fig. 1. Hence, detection becomes easier and faster. Differences in the classification results between the human operator and the automatic recognition were significant for the detection rate of CAEPs at a 20-dB SL stimulus level, based the one sided McNemar-Test.

Therefore for EEG measurements with a stimulus level close to the hearing threshold, the automatic sequential detector achieves a much sharper discrimination between measurements with and without CAEP than the human operator. The latter yields a 6% higher false positive rate for measurements without CAEP and a performance 15% below the false negative rate of the automatic detector for stimuli 20 dB above the known threshold of hearing. Concerning the decision time shown in Fig. 7(b), longer measurements in case of the human operator are due to hesitation in cases where the decision is difficult. The automatic sequential detector outperforms the human operator for most stimulus levels. The longest measurement times are necessary for the case of no acoustical stimulation, i.e., the absence of a CAEP from the EEG.

The CAEP classification by the human expert was repeated six months after the first assessment with the same data and method. From the total of 704 measurements 570 (81%) were identically classified in both situations. In more clinically oriented studies on assessment of hearing via CAEP measurements by other human operators "CAEP detection thresholds" are reported to vary between 16 dB [38] and 27 dB [39] above subjective hearing thresholds. The repeated decisions of the human operator consulted in this study were well within these limits.

Although the results obtained by the human operator should be representative for the state of the art in CAEP classification for hearing assessment in clinical routine, differences between the procedure in our study and the standard in clinical routine must be noted. First, the human operator in our study was informed about the *a priori* probability of 50% for a CAEP. As this information is unknown in clinical routine, generally the clinical assessment might be worse compared to decision rate in the study. Secondly, the CAEP analysis in clinical routine is usually based upon a set of CAEP measurements at different stimulus levels. This permits to see the dependency of amplitudes and latencies on stimulus levels, which can enhance the correct classification rates in the routine compared to our study.

It should further be noted that the human operator uses the accumulated sweep information differently from the sequential detection. His decision is based on linearly averaged sweeps, while the proposed detection method is based on binary non-linear classifications of single sweeps. For the later, the SNR of the data has to permit retrieval of sufficient information from single sweep data, as otherwise such classification would not be optimal. In the application to CAEPs evidently enough information is contained in single sweeps for decision making even at low stimulus levels.

In summary, the performance of the sequential classification can be regarded at least as good as the human operator. The rate of correct decisions of the automatic detector is identical to or higher than those attained by the human expert. Therefore, the proposed automatic sequential detector appears to give a superior performance over an experienced human operator, yielding a more accurate decision in shorter time. The better discrimination for stimulation levels close to the hearing threshold is particularly important, as it enables to determine the objective audiogram more accurately.

## V. CONCLUSION

We have introduced an automatic detection method for CAEPs, based on a DWT for feature extraction, a feature classification using a NN, and a decision mechanism based on a sequential statistical assessment of the NN's output. Although the presented feature extraction by a dyadic DWT may not optimally represent the CAEP, and despite the very high type-I-error inherent in the NN classification, the final detector stage with the sequential test is able to produce satisfying results. Applied to a study, the comparison of our method to results obtained by an experienced human operator was favorable in reducing both decision errors and measurement time, particularly when operating with stimulus levels in the proximity of the hearing threshold. This underlines the suitability of the proposed approach, and the prospect to make clinical audiometric investigations via CAEP more accurate and faster.

Although there exists a considerable quantity of contributions dealing with the signal processing aspect for auditory evoked potentials, only few of these methods have been established in clinical routine since they are not adjusted to the clinical requirements such as robustness for noisy data, restrictions on measurement duration, and exactly known rates of false positives. Clinically, the detection of a CAEP when no CAEP is present (type-I-error) is the most problematic case since it will result in failure to give, for example, a child the help it needs by erroneously judging hearing to be present, at least in CAEP terms. Therefore, the appeal of the proposed automatic detector lies in its known statistical confidence, together with its good discrimination and the reduction in measurement time.

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