

# MATCHING PURSUITS BASED ON PERCEPTUAL DISTORTION MINIMIZATION FOR SINUSOIDAL AUDIO MODELLING

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**Abstract:** In this paper, we propose an improved sinusoidal audio modeling method for perceptual matching pursuits driven by a perceptual distortion measure. A linear model derived from the effective signal processing in the ear is used for computing the perceptual distortion measure. This distortion measure allows us to select the most perceptually meaningful sinusoid at each iteration of the pursuit. Furthermore, the distortion measure can be used to define a psychoacoustic stopping criterion for the matching pursuit algorithm. The proposed sinusoidal modeling method is designed to be used in sinusoidal audio coding. Our method provides significant advantages regarding previous works because it achieves a better separation between tones and noise, as can be seen in results.

## 1 INTRODUCTION

Sinusoidal audio coding is a promising technique for audio signal characterization, compression and modification (MPEG, 2003)(Goodwin, 1998). Sinusoidal modelling is able to parameterize most of the audio signal energy in a small set of parameters because audio signals are often strongly tonal.

The classical sinusoidal model (McAulay and Quatieri, 1986) comprises an analysis-synthesis framework that represents a signal  $x[n]$  as the sum of a set of  $K$  sinusoids with time-varying frequencies, phases, and amplitudes:

$$x[n] \approx \hat{x}[n] = \sum_{k=1}^K A_k[n] \cdot \cos\left(\omega_k[n] \cdot n + \phi_k[n]\right) \quad (1)$$

where  $A_k[n]$ ,  $\omega_k[n]$  and  $\phi_k[n]$  represent the time-varying amplitude, frequency and phase of the  $k$ -th sinusoid.

Since an audio signal is typically non-stationary, it must be properly segmented in such a way that the sinusoidal parameters (amplitude, frequency and phase) change very little along each analysis audio frame. Assuming that parameters of expression (1) do not change considerably along the analysis frame, they can be made constant within a frame ( $A_k, \omega_k, \phi_k$ ) and the signal can be reconstructed from these sinusoidal parameters on a frame-by-frame basis.

A large number of methods has been proposed in the literature for estimating the parameters of the sinusoidal model. This is typically accomplished by peak picking the Short-Time Fourier Transform (STFT). Usually, analysis-by-synthesis is used in order to verify the detection of every spectral peak. In this paper we focus on the matching pursuit algorithm (Mallat and Zhang, 1993), that is a particular analysis-by-synthesis method.

Matching pursuit is an iterative algorithm that offers a sub-optimal solution for decomposing a signal  $\mathbf{x}$  in terms of unit-norm expansion functions  $\mathbf{g}_m$  chosen from an overcomplete dictionary  $D$ . At the first iteration, the function (or atom)  $\mathbf{g}_m$  which gives the largest inner product with the analyzed signal  $\mathbf{x}$  is chosen. The contribution of this function is then subtracted from the signal and the process is repeated on the residue. At the  $i$ -th iteration, it follows:

$$\mathbf{r}^i = \begin{cases} \mathbf{x} & i = 0 \\ \mathbf{r}^{i-1} - \alpha_{m(i)} \mathbf{g}_{m(i)} & i > 0 \end{cases} \quad (2)$$

where  $\alpha_{m(i)}$  is the weight associated to the optimum atom  $\mathbf{g}_{m(i)}$  at the  $i$ -th iteration and  $\mathbf{r}^0$  is initialized to  $\mathbf{x}$ . Computing the orthogonal projections of  $\mathbf{r}^{i-1}$  on elements  $\mathbf{g}_m \in D$ , the weight associated to each dictionary element at the  $i$ -th iteration (vector  $\alpha_m^i$ ) is achieved:

$$\alpha_m^i = \frac{\langle \mathbf{r}^{i-1}, \mathbf{g}_m \rangle}{\langle \mathbf{g}_m, \mathbf{g}_m \rangle} = \frac{\langle \mathbf{r}^{i-1}, \mathbf{g}_m \rangle}{\|\mathbf{g}_m\|^2} = \langle \mathbf{r}^{i-1}, \mathbf{g}_m \rangle \quad (3)$$

The optimum atom  $\mathbf{g}_{m(i)}$  at the  $i$ -th iteration is obtained by minimizing the residual energy:

$$\mathbf{g}_{m(i)} = \arg \min_{\mathbf{g}_m \in D} \|\mathbf{r}^i\|^2 = \arg \max_{\mathbf{g}_m \in D} |\alpha_m^i| \quad (4)$$

The matching pursuit algorithm (Mallat and Zhang, 1993) is energy adaptive since the optimum atom at each iteration is chosen by minimizing the residual energy according to expression (4). However, perceptual adaptation of the pursuit is desirable in order to select the most perceptually important atom instead the most correlated one with the current residue.

The main goal of this paper is to provide a psychoacoustic based-sinusoidal audio modelling method which defines a perceptual distortion measure using a linear model of the ear (S. Van de Par and Heusdens, 2002). It must be stressed that previous studies have already presented techniques for psychoacoustic-adaptive matching pursuits (Verma and Meng, 1999)(R. Heusdens and Kleijn, 2002)(Vera-Candeas et al., 2006). However, our method defines a psychoacoustic stopping criterion for the pursuit and also achieves a better separation capability between tones and noise in sinusoidal audio coding.

## 2 PERCEPTUAL DISTORTION MEASURE

A perceptual adaptation of matching pursuits can be achieved by defining a perceptual distortion measure. We have used the perceptual distortion measure introduced in (S. Van de Par and Heusdens, 2002), which is based on the monaural masking model described in (T. Dau and Kohlrausch, 1996).

The model assumes that the auditory system can be modelled by the scheme presented in Figure 1. As can be seen, the signal is filtered by the outer and middle ear response,  $h_{om}$ , and then by an auditory filter bank consisting of gamma-tone filters,  $h_b$ . This filter bank modelizes the behavior of the basilar membrane within the inner ear. Let  $\mathbf{x}$  denote the input au-

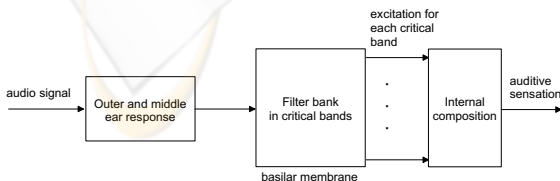


Figure 1: Model of the ear processing as a LTI system.

dio signal and  $\mathbf{d}$  the distortion signal to be computed. For each auditory band, a distortion can be measured.

This distortion is defined as the relation between the distortion energy and the signal energy at the  $b$ -th auditory band (S. Van de Par and Heusdens, 2002),

$$PDM_b(\mathbf{d}) = C_s N \frac{\|\mathbf{d}_b\|^2}{\|\mathbf{x}_b\|^2 + C_a} \quad (5)$$

where  $N$  is the signal length,  $\|\mathbf{x}_b\|^2$  the signal energy at the output of the  $b$ -th auditory filter,  $\|\mathbf{d}_b\|^2$  the distortion energy at the output of this filter,  $C_a$  the internal noise energy and  $C_s$  a constant required for calibration.

The total distortion can be computed as the sum of all distortions for each auditory filter (S. Van de Par and Heusdens, 2002), supposing that this sum corresponds to the internal composition made to conform the auditive sensation (see Figure 1). The perceptual distortion measure is defined as (S. Van de Par and Heusdens, 2002),

$$PDM(\mathbf{d}) = C_s N \sum_b PDM_b(\mathbf{d}) = C_s N \sum_b \frac{\|\mathbf{d}_b\|^2}{\|\mathbf{x}_b\|^2 + C_a} \quad (6)$$

A distortion signal is therefore audible when its PDM value is higher or equal than one.

## 3 PERCEPTUAL MATCHING PURSUITS

In order to achieve a perceptual adaptation of matching pursuits, the choice of the optimum atom has to be modified taking psychoacoustic principles into account. Several methods have been presented in the literature for such a goal (Verma and Meng, 1999) (R. Heusdens and Kleijn, 2002)(Vera-Candeas et al., 2006), all of them aiming to choose the most perceptually important atom at each iteration of the pursuit.

In standard energy-adaptive matching pursuit, the optimum atom which minimizes the energy of the residue is directly selected at each iteration. The main idea about perceptual adaptation (Verma and Meng, 1999) is to modify the weights, aiming to take perceptual principles into account. This idea can be expressed in the following way,

$$\mathbf{g}_{m(i)} = \arg \max_{\mathbf{g}_m \in D} |\alpha_m^i|_{\text{perceptual}} \quad (7)$$

The perceptual weights in equation (7) can be defined by modifying the original weights with the help of the masking threshold. In this way, Weighted Matching Pursuits (WMP) are defined in (Verma and Meng, 1999). In (R. Heusdens and Kleijn, 2002), Psychoacoustic-Adaptive Matching Pursuits (PAMP) are defined using a perceptual norm. This perceptual

norm is calculated as the integration of the ratio between the signal energy and the masking threshold in the frequency domain. It defines an inner product which facilitates the selection of the best matching dictionary element in a perceptual point of view.

However, PAMP does not define a psychoacoustic stopping criterion. The inner product does not offer any information about if a selected tone is audible or not. Furthermore, this problem can find a worse scenario, as stated in (R. Heusdens and Kleijn, 2002): PAMP can select noisy energy as a tone when zero-mean gaussian noise is present in the signal.

In our implementation, the main idea consists in modifying standard matching pursuit so as to minimize the perceptual distortion measure of the residue at each iteration of the pursuit,

$$\mathbf{g}_{m(i)} = \arg \min_{\mathbf{g}_m \in D} PDM(\mathbf{r}^i) \quad (8)$$

The optimum atom according to equation (8) can be computed applying the orthogonality property of matching pursuits. Following a matching pursuit approach, the atoms are weighted by the coefficients  $\alpha_m^i$  obtained from the correlation as is indicated in equation (3). Therefore, the residue at the  $i$ -th iteration can be decomposed as  $\mathbf{r}^{i-1} = \mathbf{r}^i + \alpha_m^i \mathbf{g}_m$ , fulfilling that  $\mathbf{r}^i$  and  $\alpha_m^i \mathbf{g}_m$  are orthogonal. Due to this orthogonality property, the perceptual distortion measure of the  $\mathbf{r}^{i-1}$  residue in a matching pursuit approach can be written as

$$PDM(\mathbf{r}^{i-1}) = PDM(\mathbf{r}^i) + PDM(\alpha_m^i \mathbf{g}_m) \quad (9)$$

As a consequence, the minimization of the perceptual distortion measure of the  $\mathbf{r}^i$  residue is the same than maximizing the perceptual distortion measure of the weighted atoms  $\alpha_m^i \mathbf{g}_m$ . Note that in a matching pursuits approach, the perceptual distortion  $PDM(\mathbf{r}^{i-1})$  is a constant at the  $i$ -iteration. Standard matching pursuit algorithm chooses at the  $i$ -th iteration the most correlated atom with the residual  $\mathbf{r}^{i-1}$  in order to minimize residual energy. Expression (9) allows us to state that choosing the weighted atom with the highest perceptual distortion measure as the optimum atom, the perceptual distortion measure of the  $\mathbf{r}^i$  residue at the  $i$ -th iteration is minimized. Perceptual matching pursuit computes the perceptual distortion measure associated to each weighted atom and selects the atom with the highest measure as the optimum atom at the  $i$ -th iteration,

$$\mathbf{g}_{m(i)} = \arg \max_{\mathbf{g}_m \in D} PDM(\alpha_m^i \mathbf{g}_m) \quad (10)$$

Distortion signals to be measured at each iteration of perceptual matching pursuit are the weighted atoms  $\alpha_m^i \mathbf{g}_m$ . The perceptual distortion measure of weighted

dictionary elements at the  $i$ -th iteration is expressed as,

$$PDM(\alpha_m^i \mathbf{g}_m) = C_s N \sum_b \frac{\|(\alpha_m^i \mathbf{g}_m)_b\|^2}{\|\mathbf{x}_b\|^2 + C_a} \quad (11)$$

The psychoacoustic stopping criterion can be directly defined in our approach. The pursuit should be halted at the iteration in which all perceptual distortions are below one. Under this condition, all remaining tones are assured to be inaudible. This condition can be expressed as:

$$PDM(\alpha_m^i \mathbf{g}_m) \leq 1, \forall \mathbf{g}_m \in D \quad (12)$$

The overcomplete dictionary  $D = \{\mathbf{g}_m[n]\}$  to be considered for sinusoidal modelling is composed of unit-norm complex exponentials.

## 4 RESULTS

First, we intend to illustrate the advantages of using  $PDM(\alpha_m^i \mathbf{g}_m)$  based on a perceptual distortion measure against the inner products  $|\alpha_m^i|_{PAMP}$  defined in (R. Heusdens and Kleijn, 2002).

Figure 2 shows the inner products  $|\alpha_m^1|_{PAMP}$  and the perceptual distortion measures  $PDM(\alpha_m^1 \mathbf{g}_m)$ , both at the initial iteration, for a windowed input signal composed of one tone plus noise. The tone power is 19 dB above the density level of noise, the tone frequency is 500Hz and the overcomplete dictionary is composed of  $M = 4096$  complex exponentials.

In this case, the PAMP approach does not select the tone correctly because medium frequency noise achieves more perceptual significance than the tone itself. As can be seen in the same figure, the PDM approach performs a right tonal extraction.

The proposed psychoacoustic stopping criterion performs correctly in this case, because after the first iteration all perceptual distortions are below 0 dB.

The better performance of our approach also happens when noise is added to a voiced speech fragment.

Figure 3 illustrates the performance of the PAMP approach when zero-mean white Gaussian noise is added to a 23-ms voiced speech fragment, being 0 dB the signal-to-noise ratio. The magnitude of all extracted tones at each iteration is drawn in circles. As can be seen, the maximum value of the perceptual inner products is just below 2 KHz, which corresponds to the most perceptually important sinusoid. This sinusoid is the first one to be extracted, giving rise to the plots in Figure 3(b). The left hand plot in Figure 3(b) also shows the magnitude and frequency of the extracted tone at the first iteration. It can be observed on the right hand plot in Figure 3(b) that perceptual distortion fall in the frequency region of the

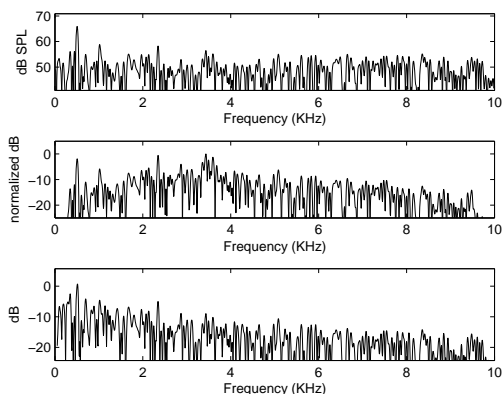


Figure 2: Perceptual inner products at the initial iteration for PAMP approach (middle plot) and perceptual distortion at the initial iteration for PDM approach (bottom plot) when the analyzed signal consists of one tone plus white gaussian noise. The tone power is 19 dB above the noise density level. The top plot shows the energy spectrum of the input signal ( $|R^0(f)|^2$ ).

extracted tone at the previous iteration. Finally, Figure 3(c) shows the residue, extracted tones and inner products at the fifth iteration. As can be seen in the left hand plot, four tones have been extracted, one of them being a high frequency tone (close to 4 KHz) that belongs to the noisy region of the signal. It can also be seen that low frequency tones still remain in the signal.

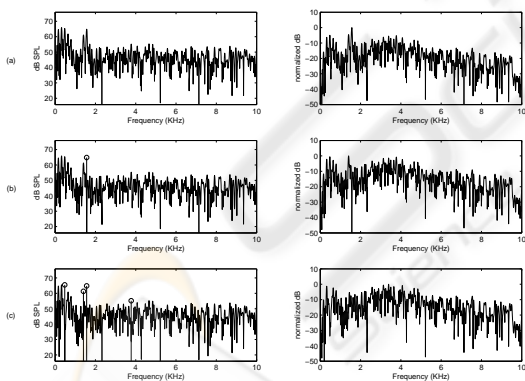


Figure 3: (a) Energy spectrum of the residue  $R^0(f)$  (energy spectrum of the input signal) and perceptual inner products for PAMP approach at the first iteration when the analyzed signal is a 23-ms voiced speech plus noise fragment, (b) Idem at the second iteration and (c) Idem at the fifth iteration.

The results obtained by our approach when using the same 23-ms voiced speech plus noise frame are somewhat different. They are shown in Figure 4. The magnitude of all extracted tones at each iteration is drawn in circles. The second column of plot (d) simply shows the extracted tones after applying the stop-

ping criterion proposed in this paper. Now, the perceptual distortion measure in the noisy region of the spectrum is lower and, at the sixth iteration, the pursuit has not yet extracted a noisy tone, as happened in the PAMP approach. All extracted atoms correspond to sinusoids in the low frequency range (below 2 KHz). Therefore, we can state that PDM-based perceptual distortion computation provides higher robustness against modelling a noisy spectrum as a tone at high frequencies than the PAMP-based approach. This modelling mistake can provoke annoying sound artifacts.

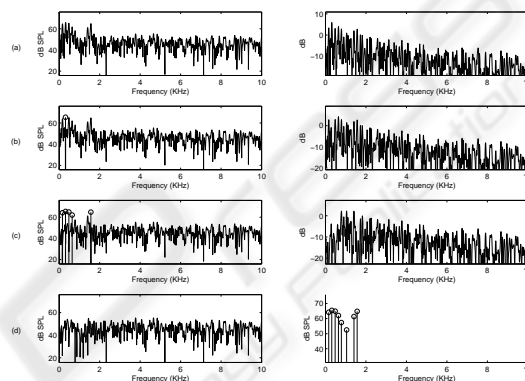


Figure 4: (a) Energy spectrum of the residue  $R^0(f)$  (energy spectrum of the input signal) and perceptual distortions for PDM approach at the first iteration when the analyzed signal is a 23-ms voiced speech plus noise fragment, (b) Idem at the second iteration, (c) Idem at the sixth iteration and (d) Residue and extracted tones at the last iteration (9-th iteration).

Finally, the operation of the psychoacoustic stopping criterion for a 23-ms voiced speech signal plus noise is shown in Figure 4. According to the proposed stopping criterion, the pursuit is halted when all perceptual distortions are below one (0 dB), which happens at the 9-th iteration (the last iteration of the pursuit in this example). Applying this criterion to our example, all perceptually meaningful tones have been extracted at the 9-th iteration. The psychoacoustic stopping criterion also help us to better discriminate between tones and noise. It must be stressed that frames where the noise power is meaningful will only have a few perceptually relevant tones. In these frames, the stopping criterion allows us to halt the algorithm at an early iteration, avoiding that the pursuit models a noisy region of the signal as a tone.

In order to compare the subjective behavior of different definitions for perceptual matching pursuits, each segment of the excerpts listed in table 1 is modelled by choosing the 25 perceptually more relevant tones. The signal to be rated corresponds to this sinusoidal part without quantization. The comparison



is based on rating the preference for the proposed PDM-based approach versus PAMP (R. Heusdens and Kleijn, 2002). CD-quality one-channel music and speech signals taken from the set of excerpts used in the MPEG standardization activities (ISO/MPEG, 2001) are chosen for testing. For comparison purposes, the analysis/synthesis is done on a frame-by-frame basis using a 50% overlap 23-ms Hanning window. A subjective listening test is performed using the double blind triple stimulus methodology, in which signal triplets OAB are presented to twelve experienced listeners. Here, O is the original signal, while A and B are the modelled signals using the PDM and PAMP approaches, respectively. The results averaged over all listeners for the 25 extracted sinusoids are shown in table 1.

Table 1: Subjective listening tests. Signals are modelled from the 25 most perceptually important sinusoids extracted according to each approach.

Excerpt	Preference (%) for PDM-MP vs. PAMP
Suzanne Vega	75
German male speech	83
English female speech	100
Harpichord	100
Castanets	58
Pitch pipe	42
Bagpipes	67
Glockenspiel	58
Plucked strings	100
Trumpet solo	67
Orchestra piece	100
Contemporary pop	100

As shown in table 1, PDM-based matching pursuit outperforms PAMP-based approach for most of the signals taken for testing. The main reason for the improvement is that PAMP sometimes extracts high frequency components representing noise, which produces annoying sound artifacts (sharp sounds). This effect is reduced with the proposed method, so that a higher audio quality is achieved. However, the improvement regarding PAMP depends on the nature of the audio signal. For those signals composed of tones and noise with meaningful energy in high frequency, such as English female speech, harpichord, plucked strings, orchestra and contemporary pop, all listeners voted in favor of our approach. For the rest of the test signals, the subjective differences are lower, because these signals have less mixed components (tones and noise) and both methods behave well.

## 5 CONCLUSIONS

This paper deals with the application of matching pursuits based on a perceptual distortion measure in order to improve sinusoidal audio modelling. As shown in the results, the proposed method achieves higher perceptual quality for the synthesized signal than the PAMP-based method when the number of frequencies to be extracted is the same. Besides, the proposed perceptual distortion measure allows us to define a perceptual stopping criterion for the pursuit. Making use of this stopping criterion, our approach has an important advantage: higher protection (sturdiness) against noise, mainly in high frequencies, which results in a better balance between tones and noise. Matching pursuit based on perceptual distortion minimization is therefore a promising technique for sinusoidal audio modelling, which allows to achieve high quality low bit-rate audio coding and has interesting applications as internet audio streaming.

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