A FAMILY OF RANDOM WAVEFORM MODELS FOR AUDIO CODING

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ABSTRACT

We study the behavior of hybrid random waveform models for audio signals, involving sparse random series of waveforms, with random coefficients. Similar approaches have been considered in the recent years. However, these do generally not rely on explicit models, and are of more "algorithmical" nature. The models we propose allow us to analyze mathematical properties of such signals and corresponding estimators, and derive estimation algorithms, which do not rely on complex optimization techniques.

1. INTRODUCTION

Recently, models for audio signal involving decompositions signal = tonal + transient + residual

have been proposed [1, 2], together with algorithms for estimating the corresponding components, with various applications. Often, the models were not specified explicitly, the proposed methods being more algorithmic in nature (see for example [3, 4, 5, 6]).

We study here more specific models, based on expansions on waveform systems. The main idea is to start with an orthonormal basis (or a frame, see [7] for the definition) of waveforms with respect to which a given component (tonal, or transient) is assumed to admit sparse expansions. Starting from an explicit model for such expansions yields estimates for the behavior of the *observed* coefficients (i.e. the coefficients computed from the realization of the signal) with respect to these bases (or frames), which we use to derive corresponding estimation algorithms.

We work here in a finite N-dimensional setting, and denote by I_N a corresponding index set. We denote by **U** and Ψ two orthonormal bases of \mathbb{C}^N , and consider the *dictionary* constructed as the union $\mathbf{U} \cup \Psi$, on which sparse signal expansions are to be seeked. The models under study are based on the following ingredients :

- Significance maps: two (small) subsets Λ and Δ of the index set I_N, modelled as random sets.
- *Coefficients*: to each δ ∈ Δ (resp. λ ∈ Λ) is associated a random variable β_δ (resp. α_λ).

The random waveform model associated with these ingredients takes the form of a linear combination of the (random) elements of Ψ and U labelled by the significance maps Λ and Δ respectively, with (random) coefficients β_{δ} and α_{λ} . Given such a signal model, we address the following main problems : from one (or several) realization(s) of the signal, assuming *sparsity* (i.e. the significance maps are small sets) and *dictionary incoherence* (i.e. the two bases are "significantly different"),

- Estimate the parameters of the model (variances of coefficients, distribution of the significance maps,...)
- Estimate the significance maps Λ and Δ .
- Estimate the two *layers* : tonal $x_{\mathbf{U}} = \sum_{\delta \in \Delta} \beta_{\delta} u_{\delta}$ and transient $x_{\Psi} = \sum_{\lambda \in \Lambda} \alpha_{\lambda} \psi_{\lambda}$.

Several approaches may be developed to solve such problems, including Bayesian MCMC type algorithms [8]. We stick here to simpler approaches, based on the behavior of *observed coefficients*, which present the advantage of lower computational load. The distribution of observed coefficients turns out to be understandable in a farily simple way, which justifies the algorithmic approach we propose. In a few words, when the layers are sparse enough in the considered bases, and if the latter are different enough from each others, the pdf of the coefficients of the signal with respect to the U and Ψ bases may be approximated as a mixture of two (or more) gaussian pdfs. Then, standard procedures may be developed for the estimation of parameters and significance maps, and therefore for the layers. Our approach differs from that of [1, 2] in that it permits the simultaneous estimation of both layers, rather than a sequential estimation.

The random waveform model and the corresponding estimation problem are the main aspects that we discuss here. The random waveform model is presented in Section 2, where the behavior of observed coefficients is also studied. The estimation algorithms and corresponding numerical results are given in Section 3, and Section 4 is devoted to an outline of possible extensions of the model, and applications.

2. HYBRID WAVEFORM MODELS

2.1. Generalities

Let \mathcal{H} denote a finite dimensional Hilbert space of dimension N, let $I = \{1, \ldots N\}$ and let $\Psi = \{\psi_n, n \in I\}$ and $\mathbf{U} = \{u_n, n \in I\}$ be two orthonormal bases of \mathcal{H} . We denote by $\mathcal{D} = \Psi \cup \mathbf{U}$ the dictionary made as the union of these two bases. \mathcal{D} is clearly (over)complete in \mathcal{H} , and any $x \in \mathcal{H}$ admits an infinite number of expansions in the form $x = \sum_{n=1}^{N} \alpha_n \psi_n + \sum_{m=1}^{N} \beta_m u_m$, where $\alpha_n, \beta_m \in \mathbb{C}$. We are interested in *sparse signals*, i.e. signals $x \in \mathcal{H}$ that may be written as

$$x = \sum_{\lambda \in \Lambda} \alpha_{\lambda} \psi_{\lambda} + \sum_{\delta \in \Delta} \beta_{\delta} u_{\delta} + r , \qquad (1)$$

where Λ, Δ are small subsets of the index set $\{1, 2, \dots N\}$, and $r \in \mathcal{H}$ is a small (possibly vanishing) residual.

Given such a sparse signal, the non-uniqueness of its expansion with respect to the dictionary makes it difficult to identify unambiguously the expansion (1), and one must rely to additional arguments to impose sparsity constraints. Greedy algorithms provide a way to

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enforce such constraints. In such situations, it may be shown that if the two bases Ψ and U are sufficiently *incoherent*, the expansion may be recovered if the sets Λ and Δ are small enough.

In the numerical examples presented in this work, we limit ourselves to two different pair of orthonormal bases : U will be a local trigonometric basis (tuned in such a way to achieve good frequency resolution), and Ψ will be either a wavelet basis, or a local cosine basis with good time resolution. In both cases, the index sets will actually be two-dimensional (a time index and a frequency index), and we shall write them as such when necessary. Other choices for the bases are possible, as well as extensions to frames (not considered here).

2.2. Random hybrid models

We now introduce an explicit *model* for the sparse signal in (1). The ingredients of such models are essentially twofold : a model for the sets Λ and Δ , hereafter termed *significance maps*, and, given these significance maps Λ and Δ , a model for the coefficients $\alpha_{\lambda}, \lambda \in \Lambda$ and $\beta_{\delta}, \delta \in \Delta$.

Definition 1 Given two orthonormal bases in \mathcal{H} as above, a corresponding random hybrid model is characterized by

i. A discrete probability model for the significance maps. The corresponding probability measures for the (random) maps Λ and Δ will be denoted by P_Λ and P_Δ, and the expectations by E_Λ and E_Δ.
ii. A probabilistic model for the coefficients α_λ, λ ∈ Λ and β_δ, δ ∈ Δ, conditional to the significance maps. The corresponding probability measure and expectation will be denoted by P₀ and E₀.

The global probability measure and expectation will be denoted by \mathbb{P} *and* \mathbb{E} .

The simplest possible model for the significance maps is the *Bernoulli* model : given a fixed *membership probability* p, the index values $n \in \{1, \ldots N\}$ are iid, and belong to Λ with probability p and to $\overline{\Lambda}$ (the complementary set) with probability 1 - p. The membership probability for Δ will be denoted by \tilde{p} . More sophisticated models for the significance maps involve correlations between elements of the significance maps.

The simplest instance for coefficient models, to which we shall stick here, assume that significant coefficients are i.i.d. $\mathcal{N}(0,\sigma^2)$ random variables, in other words their pdf (conditional to Λ) reads

$$\rho_{\alpha_n}(z|\Lambda) = \frac{1}{\sigma\sqrt{2\pi}} e^{-z^2/2\sigma^2} \,.$$

The residual will be modeled here as a Gaussian white noise, with variance σ_0^2 .

Remark 1 Since *n* is actually a time-frequency index, an obvious (and fairly realistic) modification consists in taking frequency dependent values for the variances of α and β coefficients, and coloured stationary noise for the residual.

2.3. Behavior of the observed coefficients

In practice, the only available observation is the signal x. We choose to exploit the observed coefficients

$$a_n = \langle x, \psi_n \rangle$$
, $b_m = \langle x, u_m \rangle$ (2)

for estimating a sparse expansion. As a first step, let us start by studying the distribution of these observed coefficients, *conditional to the significance maps*. For this, we introduce the indicator random

variables $X_n^\Lambda=1$ if $n\in\Lambda$ and 0 otherwise, and similarly for $X_n^\Lambda.$ Then,

$$a_n = \langle x, \psi_n \rangle = \alpha_n X_n^{\Lambda} + \sum_{m=1}^N \beta_m X_m^{\Delta} \langle u_m, \psi_n \rangle$$
(3)

$$b_n = \langle x, u_n \rangle = \beta_n X_n^{\Delta} + \sum_{m=1}^N \alpha_m X_m^{\Lambda} \langle \psi_m, u_n \rangle \quad (4)$$

and one can state

Theorem 1 Conditional to the significance maps, the *a* and *b* coefficients are zero mean normal random variables, with covariance matrices $C_{k\ell}^{A} = \mathbb{E}_{0}\{a_{k}\overline{a}_{\ell}\}, C_{k\ell}^{A} = \mathbb{E}_{0}\{b_{k}\overline{b}_{\ell}\}$

$$\begin{aligned} \mathcal{C}_{k\ell}^{\Lambda} &= (\sigma^2 X_k^{\Lambda} + \sigma_0^2) \delta_{k\ell} + \tilde{\sigma}^2 \sum_{m=1}^N X_m^{\Delta} \langle \psi_k, u_m \rangle \langle u_m, \psi_\ell \rangle , \\ \mathcal{C}_{k\ell}^{\Delta} &= (\tilde{\sigma}^2 X_k^{\Delta} + \sigma_0^2) \delta_{k\ell} + \sigma^2 \sum_{m=1}^N X_m^{\Lambda} \langle u_k, \psi_m \rangle \langle \psi_m, u_\ell \rangle . \end{aligned}$$

In particular, the diagonal terms read

$$\mathbb{E}_0\left\{\left|a_k\right|^2\right\} = \sigma^2 X_k^{\Lambda} + \tilde{\sigma}^2 \sum_{\delta \in \Delta} \left|\langle\psi_k, u_\delta\rangle\right|^2 + \sigma_0^2 \,. \tag{5}$$

Hence, the a (resp. b) coefficients are distributed according to a (random) mixture of (several) normally distributed zero-mean random variables. The distribution of these is governed by the cross term in the right hand side of the covariance coefficients in Theorem 1. Focusing on the diagonal terms of the covariance matrix, let us introduce the following quantities

Definition 2 Let Δ and Λ be two subsets of the index set I_N . For $n \in I_N$, the Parseval weights are defined by

$$\tilde{p}_n(\Delta) = \sum_{\delta \in \Delta} |\langle \psi_n, u_\delta \rangle|^2 , \ p_n(\Lambda) = \sum_{\lambda \in \Lambda} |\langle u_n, \psi_\lambda \rangle|^2 .$$
(6)

It follows from Parseval's formula that for all n and Δ (resp. Λ), $0 \leq \tilde{p}_n(\Delta) \leq 1$ (resp. $0 \leq p_n(\Lambda) \leq 1$). The following elementary result provides an estimate for the order of magnitude of the Parseval weights :

$$\mathbb{E}_{\Lambda} \{ p_n(\Lambda) \} = p ; \quad \mathbb{E}_{\Delta} \{ \tilde{p}_n(\Delta) \} = \tilde{p} .$$
(7)

Therefore, one has

$$\mathbb{E}_0\left\{\left|a_k\right|^2\right\} = \sigma^2 X_k^{\Lambda} + \tilde{\sigma}^2 p_k(\Delta) + \sigma_0^2$$

Taking into account the distribution of Δ leads to this simple consideration on the behavior of observed coefficients : if the distribution of the Parseval weights is peaked near the origin, then the coefficients a_k will be accurately described as a mixture of two Gaussians. Otherwise, more Gaussians will have to be taken into account.

Remark 2 The distribution of the Parseval weights is however difficult to study practically, as it depends on both the significance maps and the incoherence of the dictionary \mathcal{D} . In a few words : the sparser the significance maps, the smaller the Parseval weights; and the less coherent the dictionary, the smaller the Parseval weights.

The spectrum of the covariance matrix is of interest too. The following estimate results from a direct application of Gershgorin's disk theorem.

Theorem 2 Let z be an eigenvalue of the covariance matrix C^{Λ} . Then there exists $k \in I_N$ such that

$$\left|z - \left(\sigma^2 X_k^{\Lambda} + \tilde{\sigma}^2 p_k(\Delta) + \sigma_0^2\right)\right| \le \tilde{\sigma}^2 \epsilon_k(\Delta) , \qquad (8)$$

where

$$\epsilon_k(\Delta) = \sum_{\ell \neq k} \left| \sum_{\delta \in \Delta} \langle \psi_\ell, u_\delta \rangle \langle u_\delta, \psi_k \rangle \right| \,. \tag{9}$$

Hence, if for all k one has that $p_k(\Delta) > \epsilon_k(\Delta)$, the covariance matrix is non-degenerate. However, this is not the general situation (in particular for small noise variance σ_0), and the vector of a_k coefficients is sometimes a degenerate Gaussian random vector, whose distribution may nevertheless be characterized by its characteristic function.

2.4. The case of the Bernoulli model

Assume that the points of the index set are iid. Then the probability distribution of the significance map is given by

$$\mathbb{P}\{\Delta\} = \tilde{p}^{|\Delta|} (1-\tilde{p})^{N-|\Delta|}, \quad \mathbb{P}\{\Lambda\} = p^{|\Lambda|} (1-p)^{N-|\Lambda|},$$

the coefficients marginal distribution taking the simple form

$$\rho_{a_n}(\xi) = (1-p) \sum_{\Delta} \mathbb{P}\{\Delta\} \frac{\exp\{-\xi^2/2(\tilde{\sigma}^2 \tilde{p}_n(\Delta) + \sigma_0^2)\}}{\sqrt{2\pi(\tilde{\sigma}^2 p_n(\Delta) + \sigma_0^2)}} \\
+ p \sum_{\Delta} \mathbb{P}\{\Delta\} \frac{\exp\{-\xi^2/2(\sigma^2 + \tilde{\sigma}^2 \tilde{p}_n(\Delta) + \sigma_0^2)\}}{\sqrt{2\pi(\sigma^2 + \tilde{\sigma}^2 p_n(\Delta) + \sigma_0^2)}}$$
(10)

The distribution of the coefficients is thus a mixture of two Gaussian mixtures, whose behavior is again governed by the Parseval weights. Assume for a while that the distribution of the random variables $\tilde{p}_n(\Delta)$ is essentially concentrated near a small value, say the membership probability \tilde{p} (see (7)). Then the two Gaussian mixtures are zero-mean, and possess significantly different variances. In such situations, it makes sense to try to separate them, in order to estimate those index values *n* that belong to the significance map.

2.5. Structured models

Unlike the Bernoulli model, structured significance maps models involve correlations between significance map elements. For example, assuming \mathbf{U} is an orthonormal basis of time-frequency waveforms, correlations may be introduced between successive time indices, to model time persistence properties of the corresponding (tonal) layer. Similarly, correlations between frequencies may be introduced to model signal components with short duraction, such as transients (frequency persistence). In such situations, the marginal pdf of observed coefficients is still as in (10), but the probabilities are not as simple as before.

Interestingly enough, due to the decorrelation of the α and β coefficients, the correlations in significance maps do not show up in the second order moments of the observed coefficients a and b, i.e. in matrices C^{Λ} and C^{Δ} . For instance, neither $\mathbb{E}_{\Lambda} \{ C_{k\ell}^{\Lambda} \}$ nor $\mathbb{E}_{\Delta} \{ C_{k\ell}^{\Lambda} \}$ involve the correlation functions of the significance maps $\Gamma_{k\ell}^{\Lambda} := \mathbb{E}_{\Lambda} \{ X_k^{\Lambda} X_{\ell}^{\Lambda} \}$ or $\Gamma_{k\ell}^{\Delta} := \mathbb{E}_{\Lambda} \{ X_k^{\Lambda} X_{\ell}^{\Lambda} \}$. Significance maps correlations manifest themselves in higher order moments, for example 4th order moment. Indeed, according to Isserlis' formula

$$\mathbb{E}_0\left\{a_k a_\ell a_m a_n\right\} = \mathcal{C}_{k\ell}^{\Lambda} \mathcal{C}_{mn}^{\Lambda} + \mathcal{C}_{km}^{\Lambda} \mathcal{C}_{n\ell}^{\Lambda} + \mathcal{C}_{kn}^{\Lambda} \mathcal{C}_{m\ell}^{\Lambda}$$

and the expectations \mathbb{E}_{Λ} and \mathbb{E}_{Δ} of the latter quantities do involve the significance maps correlations $\Gamma^{\Lambda}_{k\ell}$ and $\Gamma^{\Delta}_{k\ell}$. However, these expressions can hardly be exploited in estimation algorithms.

3. HYBRID MODEL ESTIMATION

From now on, we limit ourselves to the simpler, Bernoulli model. We propose a new family of "three-steps" procedures for the estimation and separation of the two layers above, in the framework of the Bernoulli model. In the numerical illustrations shown below, we shall be using a pair of orthonormal bases.

3.1. Parameter and significance map estimation

In a first step, the parameters have to be estimated. For that, the empirical distribution of the *a* and *b* coefficients is compared to a Gaussian mixture model, whose parameters are estimated using EM algorithm. When the signal admits a sparse enough expansion, i.e. when the Parseval weights are small enough, a mixture of two Gaussians may be recovered in that way. Otherwise, an iterative EM algorithm may be used to recover a more complex Gaussian mixture, further identified with a mixture of two Gaussian mixtures. In any case, EM provides estimates for the variances $\sigma^2 + \tilde{\sigma}^2 \tilde{p}_n(\Delta)$ and $\tilde{\sigma}^2 \tilde{p}_n(\Delta)$ for the a_n coefficients in (10) (and similar quantities for the b_n coefficients), as well as membership probabilities p and \tilde{p} .

Once the parameters have been estimated, it may be shown that maximum likelihood estimation of the significance maps Λ and Δ amounts to simple thresholding strategies, the threshold being set to the value for which the two pdfs intersect.

3.2. Coefficient estimation and multilayer signal decomposition

The significance maps estimation actually amounts to a dimension reduction. Given a signal x, and the estimates $\widehat{\Delta}$ and $\widehat{\Lambda}$ for Δ and Λ , let $\mathcal{H}_{\widehat{D}}$ denote the subspace of \mathcal{H} spanned by the subdictionary $\widehat{\mathcal{D}} = \{\psi_{\lambda}, \lambda \in \widehat{\Lambda}\} \cup \{u_{\delta}, \delta \in \widehat{\Delta}\}$. The estimates for the U and Ψ layers of x using the reduced dictionary are obtained from its orthogonal projection onto $\mathcal{H}_{\widehat{D}}$: the signal approximation is $\hat{x} = \hat{x}_{U} + \hat{x}_{\Psi}$, with

$$\hat{x}_{\mathbf{U}} = \sum_{\delta \in \hat{\Delta}} \hat{\beta}_{\delta} u_{\delta} , \quad \hat{x}_{\Psi} = \sum_{\lambda \in \hat{\Lambda}} \hat{\alpha}_{\lambda} \psi_{\lambda} .$$
(11)

The coefficients $\hat{\alpha}$ and $\hat{\beta}$ are obtained via

$$\left(\hat{\alpha}_{1},...\hat{\alpha}_{|\hat{\Lambda}|},\hat{\beta}_{1},...\hat{\beta}_{|\hat{\Delta}|}\right)^{T} = \mathbf{G}^{-1}\left(a_{1},...a_{|\hat{\Lambda}|},b_{1},...b_{|\hat{\Delta}|}\right)^{T}, (12)$$

G being the Gram matrix of the dictionary. Therefore, the estimation of the two layers simply amounts to a matrix inversion, of size $|\widehat{\Lambda}| + |\widehat{\Delta}|$, which is a reasonable task when the significance maps are sparse enough.

Remark 3 When the significance maps are too large, so that inverting the Gram matrix becomes computationally expensive, an alternative is provided by Wiener-type estimation [9] : estimates $\hat{\alpha}_{\lambda}$ and $\hat{\beta}_{\delta}$ are obtained by a suitable weighting of the observed coefficients :

$$\hat{\alpha}_{\lambda} = \frac{\sigma^2}{\sigma^2 + \tilde{\sigma}^2 p_{\lambda}(\Delta) + \sigma_0^2} a_{\lambda}$$

The estimates obtained are poorer estimates, but easier to compute.

3.3. Numerical illustrations

The approach proposed in this paper was implemented using the MATLAB scientific computing environment, exploiting an MDCT extracted from the Stanford WAVELAB toolbox¹.

¹See http://www-stat.stanford.edu/~wavelab

We illustrate our approach by an example of multilayered decomposition of xilophone signal (65536 samples at sampling rate $\eta = 44.1kHz$, i.e. about 1.5 sec of sound), expanded onto a dictionary constructed as the union of two MDCT bases. The two bases were given windows of respective widths 2048 samples (about 46 msec.) and 128 samples (about 3 msec). The algorithm selected $|\hat{\Delta}| = 138 \text{ tonal basis functions for the tonal layer, and } |\hat{\Lambda}| = 976$ *transient* basis functions for the transient layer, i.e. all together less than 1.7% of the number of samples. The complete decomposition took about 3 mn on a 2.6GHz linux PC with 512 Mo RAM.

We display in **Fig. 1** the estimated **U** component (termed "tonal layer") and the Ψ component (transient layer). As may be seen, the separation is quite neat, the attacks have been smoothed out in the tonal part, and the transient part is sharply peaked as expected. Even though the result is not perfect, it is nevertheless of very good quality, given the small nomber of coefficients taken into account for the reconstruction.



Fig. 1. Xilophone signal. Top : original; bottom : tonal layer and transient layer

4. CONCLUSIONS AND PERSPECTIVES

The random waveform models described in this note provide an interesting and valuable approach for modeling audio signals featuring different components such as tonals, transients,... We have presented here a simple estimation algorithm, that does not rely on complex optimization techniques, and does therefore allow computations in reasonable times. Our approach yields very sparse signal expansions, which may be of interest in various applications, in particular signal coding, following the lines of [1, 2].

Possible modifications and extensions of this approach could involve the following :

- The introduction of frequency dependent variances for α and β coefficient in the model (1). It is not difficult to see that the theoretical estimates and corresponding algorithms may be modified accordingly.
- The definition and implementation of more precise models for structured significance maps. Good candidates are provided by Markov models, and Ising-type models.

- The orthogonal projection used in (11) for estimating the sparse expansion starting from the significance map may be replaced by other approaches yielding sparser approximations (for example, an adapted version of basis pursuit [4]), and potentially better separation betwen the layers.
- In an audio coding perspective, the proposed scheme is not *scalable*, i.e. it does not allow the control of the size of the significance maps as a function of desired quality. A scalable version of this approach is currently under study.
- The adaptation of this approach to audio denoising would require further modelling of the residual r in (1). Indeed, when the "clean" signal contains essentially tonals and transients (like the xilophone signal studied here), the estimated residual will essentially contain noise. In such case, our approach may be directly used for denoising purpose. However, in some cases, the "clean" signal may contain additional wide band "stochastic like" components, which should be modelled further in order to be distinguishable from the noise.
- Extension to the multichannel situation can also be considered, which raises (among others) the following question : to which extent may the different channels share a common significance map.

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