

Binaural Scene Classification with Time-Frequency Scattering and Deep Convolutional Networks

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Abstract—The abstract goes here.

I. INTRODUCTION

Fine-grained classification of acoustic scenes is only made possible by integrating signal information over a long temporal context. Whereas a few seconds are often sufficient to recognize a speaker, a musical instrument, or a genre, it may require up to 30 seconds to disambiguate closely related acoustic scenes.

II. TIME-FREQUENCY SCATTERING

Let $\psi[t]$ an analytic band-pass filter of dimensionless frequency 1 and bandwidth $1/Q$. A filter bank of wavelets is built by dilating ψ according to a geometric sequence of scales $2^{-k_1/Q}$, where the log-frequency index k_1 takes integer values. We denote by $\psi_{k_1}[t]$ the resulting wavelets. In all subsequent experiments, ψ is designed as a Gammatone wavelet of quality factor $Q = 4$, so as to approximate the properties of the human cochlea. The wavelet transform of an audio signal $x[t]$ is obtained by convolution with all wavelets: $y_1[t, k_1] = (x * \psi_{k_1})[t]$. Applying pointwise complex modulus to y_1 yields the wavelet scalogram $x_1[t, k_1] = |y_1[t, k_1]|$, also called constant- Q transform (CQT), indexed by time t and log-frequency k_1 .

Time-frequency scattering consists in convolving $x_2[t, k_1]$ with a family of two-dimensional wavelets $\Psi_{k_2}[t, k_1]$, where the index k_2 encapsulate two values, i.e. a temporal scale α and a log-frequential scale β . The temporal scale α takes 5 values between 23 ms and 370 ms according to a geometric sequence. The log-frequential scale β takes 5 values between $1/4^{\text{th}}$ of an octave and 4 octaves, as well as their 5 "mirror" frequencies, according to a geometric sequence. In addition, the edge case $\beta = \infty$ corresponds to a log-frequential low-pass filter along 4 octaves according to all 5 temporal scales α . Finally, the edge case $(\alpha, \beta) = (\infty, 0)$ corresponds to a temporal moving average $\phi[t]$ along 370 ms without any transformation of the log-frequency axis. In sum, there are $5 \times 5 \times 2 + 5 + 1 = 56$ time-frequency scattering features for every time t and log-frequency k_1 .

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The time-frequency scattering coefficients are thus defined as

$$y_2[t, k_1, k_2] = (x_1 \overset{t, k_1}{*} \Psi_{k_2})[t, k_1] = \sum_{\tau, \kappa_1} x_1[t - \tau, k_1 - \kappa_1] \Psi_{k_2}[\tau, \kappa_1]. \quad (1)$$

For some index k_2 involving a temporal scale α , applying pointwise complex modulus to $y_2[t, k_1, k_2]$ provides translation-invariant coefficients as long as the amount of translation does not exceed α . To increase the amount of invariance to translation, we apply the low-pass filter $\phi[t]$ to the moduli, hence bringing all coefficients to the same sample rate.

$$x_2[t, k_1, k_2] = |y_2[t, k_1, k_2]| \overset{t}{*} \phi[t] \quad (2)$$

III. DEEP CONVOLUTIONAL NETWORKS

Each layer in a convolutional network typically consists in the composition of three operations: two-dimensional convolutions, application of a pointwise nonlinearity, and local pooling.

$$y_3[t, k_1, k_3] = \sum_{k_2} b_3[k_2, k_3] + \mathbf{W}_3[t, k_1, k_2, k_3] \overset{t, k_1}{*} x_2[t, k_1, k_2]. \quad (3)$$

We apply the rectified linear unit (ReLU) nonlinearity, with a rectifying slope of $\nu = 0.3$ for negative inputs.

$$y_3^+[t, k_1, k_3] = \begin{cases} \nu y_3[t, k_1, k_3] & \text{if } y_3[t, k_1, k_3] < 0 \\ y_3[t, k_1, k_3] & \text{if } y_3[t, k_1, k_3] \geq 0 \end{cases} \quad (4)$$

At the pooling step, we retain the maximal activation among neighboring units in the time-frequency domain (t, k_1) over non-overlapping rectangles of width Δt and height Δk_1 .

$$x_3[t, k_1, k_3] = \max_{\substack{0 \leq \tau < \Delta t \\ 0 \leq \kappa_1 < \Delta k_1}} \left\{ y_3^+[t - \tau, k_1 - \kappa_1, k_3] \right\} \quad (5)$$

The hidden units in x_3 are in turn fed to a second layer of convolutions, ReLU, and pooling.

$$x[t] = r \times x^{\text{L}}[t] + (1 - r) \times x^{\text{R}}[t], \quad (6)$$

where r is drawn uniformly at random in the interval $[0, 1]$.

IV. CONCLUSION

The conclusion goes here.