

CONSISTENT RECOVERY OF STIMULI ENCODED WITH A NEURAL ENSEMBLE

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ABSTRACT

We consider the problem of reconstructing finite energy stimuli from a finite number of contiguous spikes. The reconstructed signal satisfies a consistency condition: when passed through the same neuron, it triggers the same spike train as the original stimulus. The recovered stimulus has to also minimize a quadratic smoothness criterion. We show that under these conditions, the problem of recovery has a unique solution and provide an explicit reconstruction algorithm for stimuli encoded with a population of integrate-and-fire neurons. We demonstrate that the quality of reconstruction improves as the size of the population increases. Finally, we demonstrate the efficiency of our recovery method for an encoding circuit based on threshold spiking that arises in neuromorphic engineering.

Index Terms— time encoding, spiking neurons, consistent recovery.

1. INTRODUCTION

Formal spiking neuron models, such as integrate-and-fire (IAF) neurons, encode information in the time domain [1]. Assuming that the input is bandlimited with known bandwidth, a perfect recovery of the stimulus from the train of spikes is possible provided that the spike density is above the Nyquist rate [2]. These results hold for stimuli encoded with neuron population models of a wide variety of sensory stimuli including audio [3] or video [4]. More generally, Time Encoding Machines (TEMs) encode analog amplitude information in the time domain using only asynchronous circuits [2]. Time encoding has been shown to be closely related to traditional amplitude sampling. This observation has enabled the application of a large number of results obtained in irregular sampling to time encoding.

A common underlying assumption of TEM models is that the input stimulus is bandlimited with known bandwidth. Although realistic for sensory stimuli, the bandlimitedness assumption has some caveats. Bandlimited functions require an

infinite time support and real-time algorithms that operate on finite time intervals are computationally more demanding [5]. Often, a good estimate of the bandwidth is not available because of some nonlinear processing in the transduction pathway or elsewhere (e.g., contrast extraction in vision).

In this paper we investigate the problem of reconstructing stimuli from a population of spike trains on a finite time horizon. The only assumption about the input stimuli is that they have finite energy, i.e., belong to L^2 on some time interval $[0, T]$. The reconstruction problem is not one of perfect recovery. Rather signal recovery is *consistent* and satisfies an optimal smoothness criterion. The consistency condition requires that the reconstructed signal triggers exactly the same spike train when passed through the same neuron as the original stimulus. The maximal smoothness criterion ensures that the problem has a unique optimal solution. The method was introduced in [6, 7] in the context of generalized sampling.

The paper is organized as follows. A brief introduction of time encoding for bandlimited functions is provided in section 2. Section 3 formulates the problem of consistent reconstruction from a finite number of spikes and presents its solution for several types of spiking neuron models that arise in practice. Explicit reconstruction schemes are provided and examples are presented. Finally section 4 concludes our work.

2. TIME ENCODING OF BANDLIMITED STIMULI

Let Ξ denote the space of bandlimited functions with finite energy and bandwidth Ω . Let $u = u(t), t \in \mathbb{R}$, be a signal (stimulus) in Ξ . The stimulus biased by a constant background current b is fed into an ideal IAF neuron with threshold δ and integration constant κ . Let $(t_k), k \in \mathbb{Z}$, denote the output spike train of the neuron.

A complete description of the encoding circuit above is provided by the t -transform. The latter can be written as

$$\int_{t_k}^{t_{k+1}} (b + u(s)) ds = \kappa\delta$$

or in inner product form

$$\langle u, g * 1_{[t_k, t_{k+1}]} \rangle = \kappa\delta - b(t_{k+1} - t_k) := q_k, \quad (1)$$

where $g(t) = \sin(\Omega t)/\pi t, t \in \mathbb{R}$, is the impulse response of a low pass filter with bandwidth Ω . Note that from the

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spike train $(t_k), k \in \mathbb{Z}$, a series of projections $\langle u, \phi_k \rangle$ with $\phi_k = g * 1_{[t_k, t_{k+1}]}, k \in \mathbb{Z}$, can be obtained. Therefore, stimulus recovery can be readily obtained from these projections; if these projections span the whole space Ξ perfect recovery of the signal is possible. The recovery is in addition stable provided that the set $(\phi_k), k \in \mathbb{Z}$, forms a frame for Ξ [8].

Theorem 1. *The bandlimited stimulus $u = u(t), t \in \mathbb{R}$, can be perfectly recovered from the spike train $(t_k), k \in \mathbb{Z}$, if the density of the spike train D satisfies the condition $D > \Omega/\pi$. If this condition holds, the recovered signal takes the form*

$$u(t) = \sum_{k \in \mathbb{Z}} c_k \psi_k(t),$$

with $\psi_k(t) = g(t - s_k), s_k = (t_k + t_{k+1})/2$ and the coefficients $c_k = [\mathbf{c}]_k$ are given in vector form by

$$\mathbf{c} = \mathbf{G}^+ \mathbf{q},$$

where \mathbf{G}^+ denotes the pseudoinverse of \mathbf{G} , $[\mathbf{q}]_k = q_k$, and the matrix \mathbf{G} has entries $[\mathbf{G}]_{kl} = \langle \phi_k, \psi_l \rangle, k, l \in \mathbb{Z}$.

Proof: The proof is based on density results for frames of complex exponentials and frame theory. See [3] for details. \square

The key condition for perfect recovery in Theorem 1 calls for the spike density to be above a certain threshold which depends on the bandwidth of the signal. Thus, there is a deep connection between time encoding and traditional amplitude sampling. Extensions to encoding bandlimited stimuli as well as space-time (video) signals with a population of neurons in cascade with receptive fields have also been reported [3, 4].

The perfect recovery results mentioned above are based on the premise that the bandwidth of the encoded stimulus is known. In sensory systems, however, it is common to find that the bandwidth of the signal that enters the soma of the neuron is unknown. Furthermore, stimuli have limited time support and the neurons respond with a finite number of spikes.

3. RECOVERY OF FINITE-LENGTH STIMULI

Let u be a signal of finite length and energy, i.e., $u \in L^2([0, T])$. In what follows we assume that the input stimulus u is fed to a population of N neurons. Let t_k^j denote the k -th spike of the neuron j , with $k = 1, 2, \dots, n_j$, where n_j is the number of spikes that the neuron j produces, $j = 1, 2, \dots, N$. As in section 2 the spiking of the neuron can be associated with the projection (measurement) of the stimulus on a set of functions. Through the use of the t -transform we can determine both the sampling functions and the result of the projection based only on the knowledge of the spike times.

Definition 1. *A reconstruction based on the spike times $(t_k^j), j = 1, 2, \dots, N, k = 1, 2, \dots, n_j$ is called consistent provided that the reconstructed stimulus \hat{u} triggers exactly the same spike train as the original stimulus u .*

Remark 1. *The consistency condition was introduced in the context of signal independent sampling in [9] and requires that the reconstructed signal provides the same samples as the original one when sampled with the same device, i.e,*

$$\langle u, \phi_k^j \rangle = \langle \hat{u}, \phi_k^j \rangle, \quad (2)$$

for all $j = 1, 2, \dots, N$ and $k = 1, 2, \dots, n_j$. Note however that in the neural context, the sampling functions ϕ_k are signal dependent.

Since we have a finite number of spikes, the sampling functions cannot form a frame for $L^2([0, T])$ [8] and therefore perfect recovery is not possible. We seek instead a consistent reconstruction which is also optimal in the sense of a certain criterion. We choose the plausible criterion of maximum smoothness which is equivalent to minimizing $\|\hat{u}''\|^2$. If a reconstruction \hat{u} satisfies the above, it is called the *optimal consistent reconstruction* of u .

The following notation will be used throughout this section. The vector \mathbf{q} is a column vector defined as $\mathbf{q} = [\mathbf{q}^1, \dots, \mathbf{q}^N]^T$ with $\mathbf{q}^j = [q_1^j, \dots, q_{n_j-1}^j]^T, j = 1, 2, \dots, N$. The vectors $\mathbf{p}, \mathbf{r}, \mathbf{c}$ are of the same dimensions and similarly defined. The matrix \mathbf{G} is a block square matrix defined

$$\text{as } \mathbf{G} = \begin{bmatrix} \mathbf{G}^{11} & \dots & \mathbf{G}^{1N} \\ \vdots & \ddots & \vdots \\ \mathbf{G}^{N1} & \dots & \mathbf{G}^{NN} \end{bmatrix} \text{ and } \mathbf{G}^{ij} = [G_{kl}^{ij}], i, j = 1, \dots, N, k = 1, \dots, M_i - 1, l = 1, \dots, n_j - 1.$$

3.1. Representation with a Population of LIF Neurons

Consider N leaky integrate-and-fire (LIF) neurons where neuron j has threshold δ^j , bias b^j , resistance R^j and capacitance C^j . Neuron j fires a spike when it's membrane potential hits its threshold and then it resets its membrane potential to 0. The t -transform of the population can be written as

$$\int_{t_k^j}^{t_{k+1}^j} [u(s) + b^j] e^{-\frac{t_{k+1}^j - s}{R^j C^j}} ds = C^j \delta^j$$

or in inner product form as

$$\langle u, \phi_k^j \rangle = q_k^j, \quad (3)$$

with

$$\phi_k^j = e^{-\frac{t_{k+1}^j - t}{R^j C^j}} 1_{[t_k^j, t_{k+1}^j]}(t)$$

$$q_k^j = C^j \delta^j - b^j R^j C^j \left(1 - \exp\left(-\frac{t_{k+1}^j - t_k^j}{R^j C^j}\right) \right),$$

for all $j, j = 1, 2, \dots, N$ and all $k, k = 1, 2, \dots, n_j - 1$. We have the following:

Theorem 2. *Assume that at time 0 the membrane potential of all neurons is at the rest value 0. The optimal consistent*

reconstruction \hat{u} is unique and can be written as

$$\hat{u}(t) = a_0 + a_1 t + \sum_{j=1}^N \sum_{k=1}^{n_j-1} c_k^j \psi_k^j(t), \quad (4)$$

where

$$\psi_k^j(t) = \int_{t_k^j}^{t_{k+1}^j} |t-s|^3 \exp\left(-\frac{t_{k+1}^j-s}{R^j C^j}\right) ds. \quad (5)$$

The reconstruction coefficients are given in matrix form by

$$\begin{bmatrix} a_0 \\ a_1 \\ \mathbf{c} \end{bmatrix} = \begin{bmatrix} \mathbf{p} & \mathbf{r} & \mathbf{G} \\ 0 & 0 & \mathbf{p}^T \\ 0 & 0 & \mathbf{r}^T \end{bmatrix}^+ \cdot \begin{bmatrix} \mathbf{q} \\ 0 \\ 0 \end{bmatrix}, \quad (6)$$

$$p_k^j = R^j C^j \left(1 - e^{-\frac{t_{k+1}^j - t_k^j}{R^j C^j}} \right)$$

$$r_k^j = R^j C^j \left(\left(t_{k+1}^j - R^j C^j \right) - \left(t_k^j - R^j C^j \right) e^{-\frac{t_{k+1}^j - t_k^j}{R^j C^j}} \right)$$

$$G_{kl}^{ij} = \langle \phi_k^i, \psi_l^j \rangle.$$

Proof: The unique representation result of (4) together with (5) and (6) are a direct consequence of the main result of [6].

To get some intuition, note that the quadratic criterion $\|\hat{u}\|^2$ can be written in a bilinear form as $\langle \mathcal{D} * \hat{u}, \hat{u} \rangle$, where \mathcal{D} is the fourth order differential convolution kernel

$$\mathcal{D} = \frac{d^4 \delta(t)}{dt^4}.$$

Then, the reconstruction functions of (5) can be obtained as

$$\psi_k^j = \phi_k^j * f, \quad (7)$$

where $f(t) = |t|^3$, is a Green's function for \mathcal{D} , i.e., it satisfies

$$\mathcal{D} * f(t) = \delta(t).$$

Moreover, the set $\{1, t\}$ forms a basis for the kernel of the quadratic criterion and the entries of \mathbf{p} and \mathbf{r} are given by

$$p_k^j = \langle 1, \phi_k^j \rangle, \quad r_k^j = \langle t, \phi_k^j \rangle. \quad (8)$$

Now \hat{u} satisfies (3) since from (6) we have that

$$\begin{bmatrix} \mathbf{p} & \mathbf{r} & \mathbf{G} \end{bmatrix} \cdot \begin{bmatrix} a_0 \\ a_1 \\ \mathbf{c} \end{bmatrix} = \mathbf{q}. \quad (9)$$

Finally, since each neuron starts from its reset level the reconstructed signal will generate exactly the same spike times. Thus, the reconstruction is consistent. \square

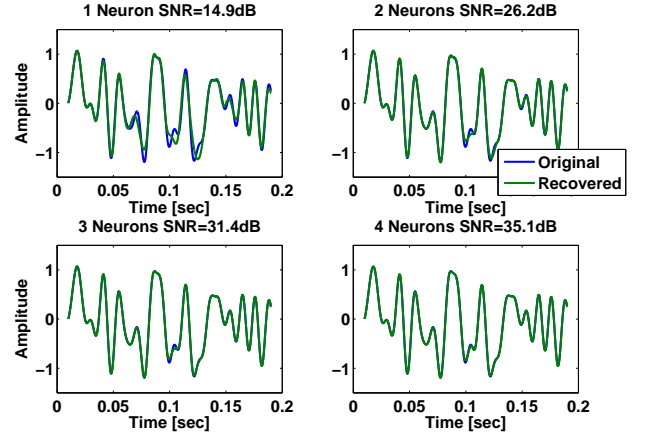


Fig. 1. Consistent reconstruction from a finite number of spikes with an ensemble of LIF neurons.

An example of the performance of the consistent recovery algorithm is shown in Fig. 1. The input stimulus was a bandlimited function with $\Omega = 2\pi 100$ rad/sec restricted to the time interval $[0, 200]$ msec. Four different LIF neurons encoded the stimulus. As it can be seen, the signal-to-noise ratio increases with the number of neurons, i.e., the number of spike trains, used for recovery. This is consistent with the evolutionary intuitive argument that increasing the number of neurons leads to an improved representation of the sensory world.

3.2. Representation with an ON-OFF AER Neuron

In this section we consider a silicon neuron model that has been used in the context of address event representation (AER) for silicon retina and related hardware applications [10]. The spiking mechanism is threshold based and simple reset mechanisms are included (see Fig. 2). The ON-OFF AER generates a spike whenever a change δ is detected. The t -transform of the ON-OFF AER neuron amounts to

$$\begin{aligned} u(t_k^1) &= u(0) + \delta \cdot \left(k - \sum_{l=1}^{n_2} 1_{\{t_l^2 < t_k^1\}} \right) := q_k^1 \\ u(t_k^2) &= u(0) - \delta \cdot \left(k - \sum_{l=1}^{n_1} 1_{\{t_l^1 < t_k^2\}} \right) := q_k^2. \end{aligned} \quad (10)$$

As in the previous examples, the above equalities can also be expressed in inner product form of (3) with $\phi_k^j(t) = \delta(t - t_k^j)$ for all $j, j = 1, 2$ and all $k, k = 1, \dots, n_j$. Note that in this case, the spiking of the silicon neuron acts as an irregular sampler on the input stimulus. We have the following theorem:

Theorem 3. The optimal consistent reconstruction \hat{u}

$$\hat{u}(t) = a_0 + a_1 t + \sum_{j=1}^2 \sum_{k=1}^{n_j} c_k^j \psi_k^j(t), \quad (11)$$

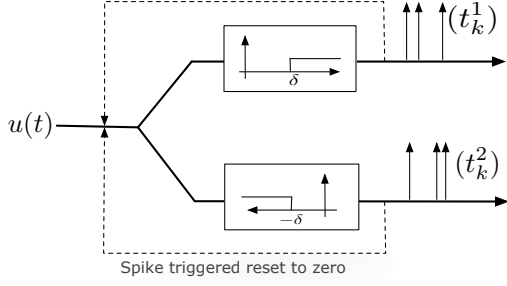


Fig. 2. Encoding with an AER Neuron.

where

$$\psi_k^j(t) = |t - t_k^j|^3 \quad (12)$$

The coefficients a_0, a_1 and \mathbf{c} are given by (6) with

$$p_k^j = 1, \quad r_k^j = t_k^j, \quad G_{kl}^{ij} = \langle \phi_k^i, \psi_l^j \rangle. \quad (13)$$

Proof: Same as in Theorem 2. Note that, as in Theorem 2, (12) can be obtained from (7) and (13) from (8). \square

In hardware implementations, the input to an ON-OFF AER neuron is usually the temporal contrast of the (positive) input photocurrent. With v the input photocurrent the temporal contrast u is defined as

$$u(t) = \frac{d \log(v(t))}{dt} = \frac{1}{v(t)} \frac{dv}{dt}.$$

In such a case, it is clear that even when the input bandwidth of the photocurrent v is known, the effective bandwidth of the actual input u to the neuron cannot be analytically estimated.

Fig. 3 shows an example. The input photocurrent v was a positive bandlimited function with $\Omega = 2\pi 40$ rad/sec. The temporal contrast u (blue line) was fed into an ON-OFF AER neuron and was recovered using (a) the consistent recovery algorithm (green line) and (b) the perfect recovery algorithm (described in section 2) with $\Omega' = \Omega$ (red line) and with $\Omega' = 5 \cdot \Omega$ (light blue line). For the consistent recovery the SNR recorded was 37.65 [dB] whereas for the perfect recovery algorithm with assumed effective bandwidth, the SNR was 8.37 [dB] and 8.38 [dB], respectively. The consistent recovery algorithm clearly outperforms the perfect recovery algorithm with assumed effective bandwidth.

4. CONCLUSIONS

We presented a new framework for the recovery of stimuli encoded into a finite number of spikes. The framework assumes that a parametric description of the encoding mechanism is available. Under the assumption that the recovered signals satisfy a quadratic smoothness criterion, we derived an algorithm to reconstruct consistent stimuli of finite energy.

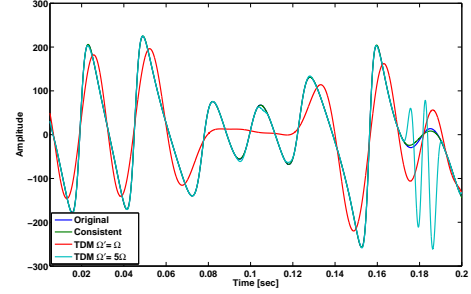


Fig. 3. Recovery from an ON-OFF AER Neuron.

We demonstrated that the algorithm performs well in practice and that it is suitable when certain signal characteristics are unknown. Our work further enhances the view of neural encoding as a set of projections of the stimulus on a family of sampling functions; it thereby builds a strong connection between representation in the spike domain and traditional sampling theory. Extensions will be presented elsewhere.

5. REFERENCES

- [1] Aurel A. Lazar, "Time Encoding with an Integrate-and-Fire Neuron with a Refractory Period," *Neurocomputing*, vol. 58-60, pp. 53–58, June 2004.
- [2] Aurel A. Lazar and László T. Tóth, "Perfect Recovery and Sensitivity Analysis of Time Encoded Bandlimited Signals," *IEEE Transactions on Circuits and Systems-I: Regular Papers*, vol. 51, no. 10, pp. 2060–2073, October 2004.
- [3] Aurel A. Lazar and Eftychios A. Pnevmatikakis, "Faithful Representation of Stimuli with a Population of Integrate-and-Fire Neurons," *Neural Computation*, vol. 20, no. 11, pp. 2715–2744, 2008.
- [4] Aurel A. Lazar and Eftychios A. Pnevmatikakis, "A Video Time Encoding Machine," in *IEEE International Conference on Image Processing*, San Diego, CA, October 12-15 2008.
- [5] Aurel A. Lazar, Ernő K. Simonyi, and László T. Tóth, "An Overcomplete Stitching Algorithm for Time Decoding Machines," *IEEE Transactions on Circuits and Systems-I: Regular Papers*, vol. 55, no. 8, September 2008.
- [6] J. Kybic, T. Blu, and M. Unser, "Generalized sampling: a variational approach. I. Theory," *IEEE Transactions on Signal Processing*, vol. 50, no. 8, pp. 1965–1976, 2002.
- [7] J. Kybic, T. Blu, and M. Unser, "Generalized sampling: a variational approach. II. Applications," *IEEE Transactions on Signal Processing*, vol. 50, no. 8, pp. 1977–1985, 2002.
- [8] Ole Christensen, *An Introduction to Frames and Riesz Bases*, Applied and Numerical Harmonic Analysis. Birkhäuser, 2003.
- [9] M. Unser and A. Aldroubi, "A general sampling theory for nonideal acquisition devices," *IEEE Transactions on Signal Processing*, vol. 42, no. 11, pp. 2915–2925, 1994.

- [10] P. Lichtsteiner, C. Posch, and T. Delbruck, "A 128×128 120 dB $15 \mu\text{s}$ Latency Asynchronous Temporal Contrast Vision Sensor," *IEEE Journal of Solid-State Circuits*, vol. 43, no. 2, pp. 566–576, 2008.