

# Comparison Between Two Spatio-Temporal Organization Maps for Speech Recognition

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**Abstract.** In this paper, we compare two models biologically inspired and gathering spatio-temporal data coding, representation and processing. These models are based on Self-Organizing Map (SOM) yielding to a Spatio-Temporel Organization Map (STOM). More precisely, the map is trained using two different spatio-temporal algorithms taking their roots in biological researches: The ST-Kohonen and the Time-Organized Map (TOM). These algorithms use two kinds of spatio-temporal data coding. The first one is based on the domain of complex numbers, while the second is based on the ISI (Inter Spike Interval). STOM is experimented in the field of speech recognition in order to evaluate its performance for such time variable application and to prove that biological models are capable of giving good results as stochastic and hybrid ones.

## 1 Introduction

Spatio-temporal classical connectionist networks comprise an important class of neural networks that can deal with patterns distributed both in time and space. In the case of Automatic Speech Recognition (ASR), this class of models have been shown to yield good performance (sometimes better than Hidden Markov Models) on short isolated speech units. By their recurrent aspect and their implicit or explicit temporal memory they can perform some kind of integration over time. Yet till now spatio-temporal biologically inspired models are the most less used for ASR. This class of models is very relevant to be exanimate because the advance realized in neurophysiologic researches have yield to the emergence of neuromimetic models especially for explicitly processing of temporal information and we know that in ASR, there is a time dimension or a sequential dimension which is highly variable and difficult to handle directly in ANNs. These models present an alternative approach to ASR which might in the long term help to overcome restrictions of current speech recognition technology with regard to noise tolerance or speaker independence. Some of these biological models have demonstrated good performance for ASR, we can cite the work of Bérroule [20 ] concerning dynamic propagation or Durand [19 ] for super units map. In this paper we present STOM map extending the SOM map from the

processing of purely spatial signals to the processing of spatio-temporal signals using biologically inspired approaches and algorithms. Moreover, STOM represents the time dimension, depending on the parameters used, either as the level of weight or the map. The use of SOM in this paper is coming from the conviction of universal auto-organization concept covering the major part of human brain [11]. This concept could be able to explain all experimental findings concerning the plasticity of cortical topographies. A possible candidate for universal self-organization is the SOM map. STOM map will be trained using two different spatio-temporal algorithms. The first one is the ST-Kohonen algorithm proposed by Mozayyani [13]. It is an algorithm using spatio-temporal inputs which are represented by a spatio-temporal coding approach proposed by Vaucher [12]. The latter uses the domain of complex numbers to encode inputs. The choice of this domain derives from the fact that it is the only domain supplying two degree of freedom allowing to represent the two correlated data. The second algorithm is TOM (Time Organized Map) [11]. Its main additional idea comparing to SOM is the functionally reasonable transfer of temporal signal distances into spatial signals distances in topographic neural representations. This is achieved by neural dynamics of propagating waves. The inputs processed by this algorithm are encoded using an approach based on the ISI (Inter Spike Interval)[2],[16]. Each input is presented as a pair containing the spatial information of the stimuli, and the temporal distance that separate it from its successor.

STOM is experimented in the domain of speech recognition of isolated digits. Section two of this paper presents the encoding approach used for STOM inputs, while section three is devoted to describe ST-Kohonen training model and propose also an extension of TOM for two dimensions. In section four, we show the application of the two encoding approaches to the analysis of acoustic signal and we compare experimental results between the two models obtained in the domain of speech recognition. Section five concludes the paper and presents some perspectives for the model.

## 2 The Spatio-Temporal Encoding Approaches

### 2.1 The Complex Approach

The spatio-temporal technique takes its roots from the work undertaken by neurobiologists to model passive electric properties of the dendrites trees [15], in particular Rall's work [17]. It is based on a particular modeling of the Post-Synaptic Potentials (PSP) and of their mix. According to the formalization made by Agmon-Snir [1], which characterizes one  $PSP(t)$  by its moments of order  $k$ , only the first two moment are kept, each one of them being respectively associated to the norm and phase of a complex number. The use of complex domain is justified by the fact that it is the only domain offering numbers having two degree of freedom. This property allows encoding the two correlated data at the same time. The Spatio-Temporal coding technique (Fig. 1.) is introduced for the aim to provide the classical artificial neurons the capacity of processing sequences in asynchronous manner, leading to the emergence of STANN [12] (Spatio-Temporal Artificial Neural Network). It consists on adding the

delay; at the level of input; to introduce temporal information. It improves the work of 'Integrate and Fire' neuron [9], [7] using the field of complex numbers. The spatio-temporal coding is done as following: Each input produces a train of impulses, for each impulse or spike  $I$  characterized by its amplitude  $\mu_I$  and the temporal delay  $d_I$  which separates the current instant of time from the time at which the impulse has been occurred, a complex number  $z_I$  is assigned.  $z_I$  Contains  $\mu_I$  as its module and  $\varphi_I$  as its phase as following:

$$z_I = \mu_I e^{i\rho_I} \quad (1)$$

$$\rho_I = \arctan(\mu_I d_I)$$

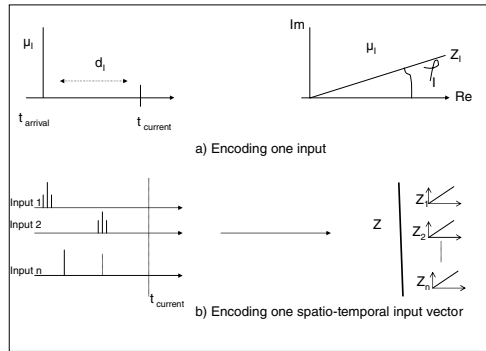
$\mu_t$  is a constant inverse to time. To incorporate the attenuation of amplitude through time, the amplitude of the input could be calculated using the following formula:

$$\mu_I(t + \Delta t) = \mu_I(t) e^{-\mu_s \Delta t} \quad (2)$$

According to Baig [14], the couple  $(\mu_s, \mu_t)$  must satisfy the following condition in order to obtain a good complex encoding.

$$\mu_s = \mu_t = 1/T_w \quad (3)$$

$T_w$  represents the temporal window used to encode all the coming inputs.



**Fig. 1.** The Spatio-Temporal Complex coding approach

## 2.2 The Temporal Coding Approach

Recent neurobiological findings experimented by Spengler et al [3] have demonstrated the importance of temporal interval between stimuli. In these experiments, tactile stimuli with different relations in space and time were applied to monkey. After months of training, the primary somatosensory cortex was mapped, showing that temporally separated stimuli (with an ISI of about 300ms) were segregated within

cortical topography, while stimuli having an ISI under 100ms are integrated. Thus the dynamic of incoming stimuli reflects the stimuli's relatedness with regard to their functional meaning. Stimulus dynamic is therefore important for the learning of topography. Accordingly, many researchers think that the design of incoming flux of  $n$  stimuli  $S_i$  could be represented as sequence of couple containing the stimulus and  $ISI_i$  interval expressing the temporal proximity of consecutive stimuli.

$$\begin{aligned} (S_i, ISI_i) \\ i = 1, \dots, n \end{aligned} \tag{4}$$

### 3 Spatio-Temporal Training

#### 3.1 The ST-Kohonen

ST-Kohonen map [14] is a SOM having complex neurons. These latter have input and weighting vectors defined in the complex domain as described in the above section. The ST-Kohonen algorithm works in the same manner as classical Kohonen one, however, the winner is chosen according to the hermitienne distance instead of the Euclidian one:

$$\delta(X, W_i) = \|X - W_i\| = \sqrt{{}^t(X - W_i)(X - W_i)} \tag{5}$$

$X$  designs the map input and  $W_i$  is the weighting vector of the neuron  $i$ . Both the input and the weight vectors are defined in the complex domain  $\mathbb{C}^p$ . The adaptation rule for ST-Kohonen is the same as the one presented in Kohonen, yet we manage complex vectors instead of real. With ST-Kohonen we made a spatio-temporal classification which is very relevant in ASR because it allows classifying speech according to its feature extractors and their temporal occurrence or sequences. However, topology obtained by the map is spatial because weights are updated according to the distance of neurons in the map and not in the input space.

#### 3.2 The Time Organized Map Algorithm (TOM)

The TOM algorithm was presented by Wiemer [11] for a better understanding of the self-organization and the geometric structure of cortical signal representations. TOM was proposed for one dimension SOM map, reader can refer to [11] for more details about the algorithm. In the present section, we present an extension of TOM for two dimensions.

High-dimensionally coded stimuli are applied to TOM two dimensions (Fig. 2.) at discrete times. They are described by a sequence  $(s_n, ISI_n)$ . The layer is composed of  $N_1^c \cdot N_2^c$  neurons ('c' for cortical). At time  $t_n$ , a stimulus  $s_n = s_A(t_n)$  is presented to the map resulting in a feedforward activation obtained by multiplying the connection matrix with the stimulus: The activity  $c_A(t_n)$  builds up from the current feedforward

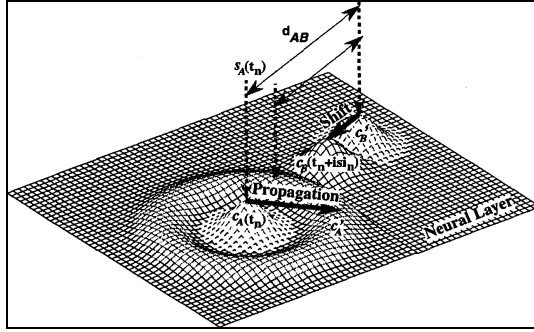


Fig. 2. Wave propagation and interaction in a two dimension TOM map

activation  $c^{ff}(t_n)$ , and the activity state of the map, that has evolved out of the earlier activity  $c(t_{n-1})$  from time  $t_{n-1}$  to  $t_n$ . The evolution of map activity between two stimulations is of wave-like type; excitation propagates into its surround. This dynamic may result from local interactions [5],[6],[8] e.g. between excitatory and inhibitory neurons. The dynamic is fundamental in the sense of a general principal of locality. In other words, 'effects propagate from point to neighboring point. In case of a monomodal and rotation symmetric of map activity, an 'elementary wave' propagates as is shown by Fig. 2. Assuming linear superposition, the dynamic corresponding to general activity patterns can be reduced to the superposition of elementary waves. As an analogue, one may think of water waves that a raindrop generates when it falls down. During the  $isi_n$ , and assuming a constant propagation speed, the wave has propagated along a certain distance reaching neurons having position that verify the following formula:

$$c'_A(n) = k_{ff}(n) \pm v.ISI_n \quad (6)$$

Where  $k_{ff}(n)$  is the neuron activated by the stimulus  $s_A(t_n)$ . The stimulus  $s_B(t_n + isi_n)$  (applied at that time), generates also a feedforward activation resulting in the activity  $c'_B$ . This activity will propagate in the map yielding to an interaction zone between the two waves. Considering this interaction, the activity of the map must be shifted to towards this zone because it consists of neurons that are excited by the two stimuli. In a neural field model, and also in biological model [4] the length of the shift depends on the distances between the wave front of the stimulus  $s_A(t_n)$ , which is  $c'_A(n)$ , and  $k_{ff}(n + isi_n)$ , the neuron activated by  $s_B(t_n + isi_n)$ . This distance could be determined using the shortest path linking the two neurons  $k_{ff}(n)$  and  $k_{ff}(n + isi_n)$ . Thus the two dimensional interaction could be reduced to one-dimensional interaction along this shortest path. This reduction allows us to apply one dimensional TOM algorithm along this path. The neurons that constitute this shortest path must have a coordinate  $(i_k, j_k)$  verifying the following formula:

$j_k = \text{round}(A i_k + b)$  and

$$\begin{aligned}
 & [ i_{k_{ff}(n)} < i_k < i_{k_{ff}(n+isi_n)} \text{ or } i_{k_{ff}(n+isi_n)} < i_k < i_{k_{ff}(n)} ] \text{ and} \\
 & [ j_{k_{ff}(n)} < j_k < j_{k_{ff}(n+isi_n)} \text{ or } j_{k_{ff}(n+isi_n)} < j_k < j_{k_{ff}(n)} ]
 \end{aligned}
 \tag{7}$$

Where:

$$A = \frac{j_{n+isi_n} - j_n}{i_{n+isi_n} - i_n}, \text{ et } b = j_n - A i_n
 \tag{8}$$

$(i_n, j_n)$  and  $(i_{n+isi_n}, j_{n+isi_n})$  are the coordinates of the neuron  $k_{ff}(n)$  and  $k_{ff}(n + isi_n)$ . Once the path is determined, the different steps proposed for TOM one dimension algorithm could be applied. These steps are as follows:

**Determination of the Shift Toward the Interaction Zone Due to the Propagation of Waves**

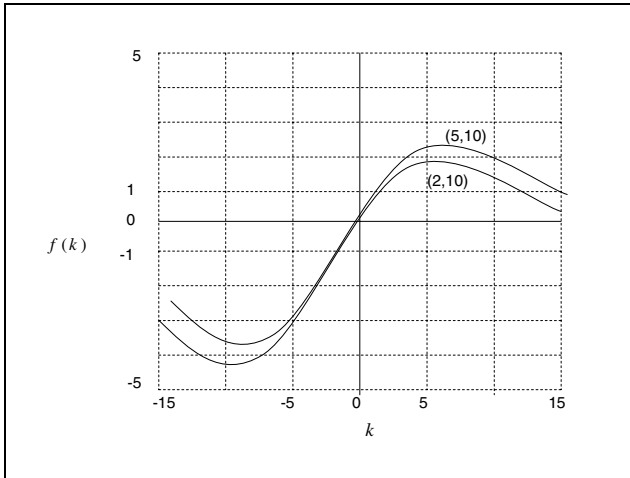
The interaction shift is calculated according to the distance that separates the current winner from the border of the propagation waves introduced by the earlier stimulus

$$\Delta_{\text{int}}(n) = f(c'_A(n) - k_{ff}(n))
 \tag{9}$$

The interaction function  $f$  expresses the distance dependency of the shift. Wiermer proposed the following function

$$f(k) = K \cdot \tanh\left(\frac{k}{K}\right) \exp\left(-\frac{k^2}{2\sigma_K^2}\right)
 \tag{10}$$

$K$  represents the strength of the interaction. The form of the interaction function is shown by (Fig. 3.) for different values of  $K$  and  $\sigma_K$ .



**Fig. 3.** The non linear interaction function for  $(K, \sigma_k) = (5,10)$  et  $(2,10)$

### Shift Due to Noise

Noise in the network is expressed by a noise term that is randomly drawn from a normal distribution with zero mean and standard deviation

$$\Delta_{noise} \rightarrow N(0, \sigma_{noise}(n)) \quad (11)$$

The standard deviation decrease monotonically in time from its initial value  $\sigma_0$  to its final value  $\sigma_f$ .

$$\sigma_{noise}(n) = \sigma_0 \left( \frac{\sigma_f}{\sigma_0} \right)^{n/n'_f} \text{ si } n \leq n'_f \quad (12)$$

And remains constant to its final value for the remaining of the steps learning.

### New Winner Position

The winner  $k_{learn}(n)$  is determined as the integer closest to the position of the maximal feedforward activation shifted by interaction and noise.

$$k_{learn}(n) = \text{round}(k_{ff} + \Delta_{int} + \Delta_{noise}) \quad (13)$$

### New Winner Adaptation Weights

Only winner weights are adapted by learning step. They are shifted toward the presented stimulus.

$$\Delta W_{k_{learn}}(n)(n) = \alpha \cdot (S_n - W_{k_{learn}}(n)(n)) \quad (14)$$

$\alpha$  is the learning rate, it is constant during all the training.

## 4 Experimentation

### 4.1 The Data Coding and Model

The map is trained and tested with isolated words of TI-Digit database. The number of speakers contributing to the application is 110 distributed as 55 men and 55 women pronouncing the eleven vocal digits (0, 1, ...,9 and 'oh'). Each speaker pronounces one digit twice: one occurrence is used for the training and the other for the test. We use the Mel Frequency Cesprtral Coefficients (MFCC) vectors [18] as feature extractor; it is composed of 12 coefficients. The speech signal were collected in quiet environment and digitized at 16 KHz. The window size for calculating each MFCC contains 512 points. Windows are overlapping and are separated with 10ms (160 points) between two consecutive MFCC. To permit the spatio-temporal coding with complex numbers, we suppose that all coefficients for one MFCC vector have the same time of occurrence. Thus, complex MFCC vectors will have the following form:

$$MFCC_{real} = \left\{ \begin{array}{c} c_1 \\ \cdot \\ \cdot \\ \cdot \\ c_{12} \end{array} \right\} \Rightarrow MFCC_{complex} = \left\{ \begin{array}{c} c_1 e^{i\rho_1} \\ \cdot \\ \cdot \\ \cdot \\ c_{12} e^{i\rho_{12}} \end{array} \right\} \quad (15)$$

The phase is calculated using the following formula:

$$\rho_i = \arctan(1 - 0.01(i - 1)) \quad (16)$$

The temporal encoding approach is straight forward. In fact each input is couple of the following form:

$$(S_n, isi_n) = \left( \left( \begin{array}{c} c_1 \\ \dots \\ c_{12} \end{array} \right), 10ms \right) \quad (17)$$

## 4.2 Experimental Results

STOM is tested by either ST-Kohonen algorithm or TOM in speech recognition. For the TOM algorithm, we have used the same parameters proposed by Wiemer which are as follows:  $v = 1, K = 5, \sigma_k = \sigma_0 = 15, \sigma_f = 0.1, \alpha = 0.01, n'_f = 0.1 \cdot 10^9$ .

These parameters are determined empirically and have demonstrated good results comparing to other values.

**Table 1.** Three protocols for digit recognition using ST-Kohonen and TOM models

Recognition rate	ST-Kohonen	TOM
Monolocator	98,56%	99,34%
Speaker independent	97,5%	98,32%
Unknown locator	96,9%	98,45%

In order to evaluate the learning and robustness capacities of the two models, we have tested them for digit recognition using three distinct protocols: the monolocator recognition, the speaker independent recognition and the unknown locator recognition. For the monolocator, we take the first occurrence of digits as learning base and the second occurrence as test base. In the Speaker independent protocol, all first occurrences of each digit form the learning base and all second occurrences of digits form the test base. For unknown locator, 28 of women and men first occurrences are taken for learning base and the 27 others for the test base. The results are reported in the above table (Table 1.). For the three protocols TOM model performs better than the spatio-temporal classification made by ST-Kohonen. Indeed, the spatio-temporal interaction between inputs can capture more the fine spatio-temporal structure inherent to speech signal. However, the main drawback of TOM is the time of convergence. In fact, the algorithm has needed  $10^9$  loops to converge. This slowness could be interpreted by the fact that TOM update only the winner by each loop yielding to



an increasing of time for the map to be organized. ST-Kohonen presents better result compared to other time-based neural networks [10]; this fact is explained by the use of coding approach gathering spatio-temporal data as inputs allowing a spatio-temporal classification.

## 5 Conclusion

The Spatio-temporal Organization Maps presented in this paper are an extension of SOM map to spatio-temporal domain. This extension is based on biologically inspired approaches related to coding or processing data. STOM is trained using two spatio-temporal processing algorithms. The results obtained by applying STOM in speech recognition are good and both algorithms are qualified for processing the fine spatio-temporal structure of speech signal. This might provide an insight into speech recognition using biologically models and could in the long run overcome the limitation of HMM-based speech technology or hybrid models regarding for example noise. Further research could be focused on testing the presented models using corpus collected in real world. It could be also possible to combine biologically model for speech perception, a model of human cochlea for example, with STOM because we think that it can be a fruitful area that needs to be explored.

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