

Imperial College  
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# Timing is everything: Sparse sampling based on time-encoding machines

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**Ack.**



Interaction  
Vincent Leung (ICL)



Joint work with  
Roxana Alexandru (ICL)

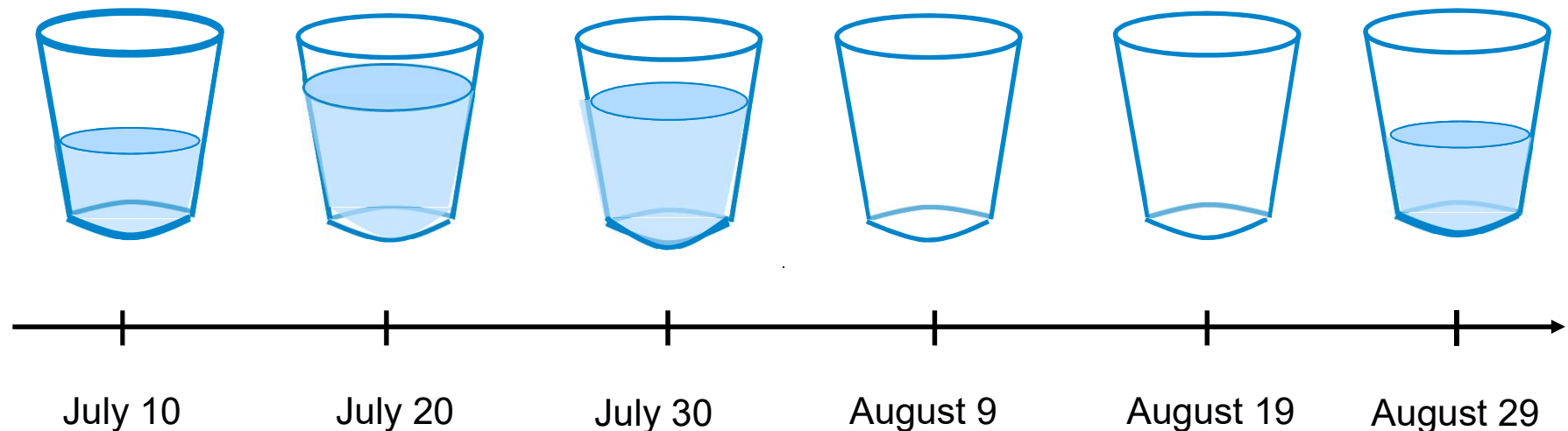
## The Problem

- Your primary school child has been assigned a nasty summer homework
- He needs to estimate rainfall over the summer break
- He is desperate because the summer vacation is at stake

## The Problem

### Approach 1

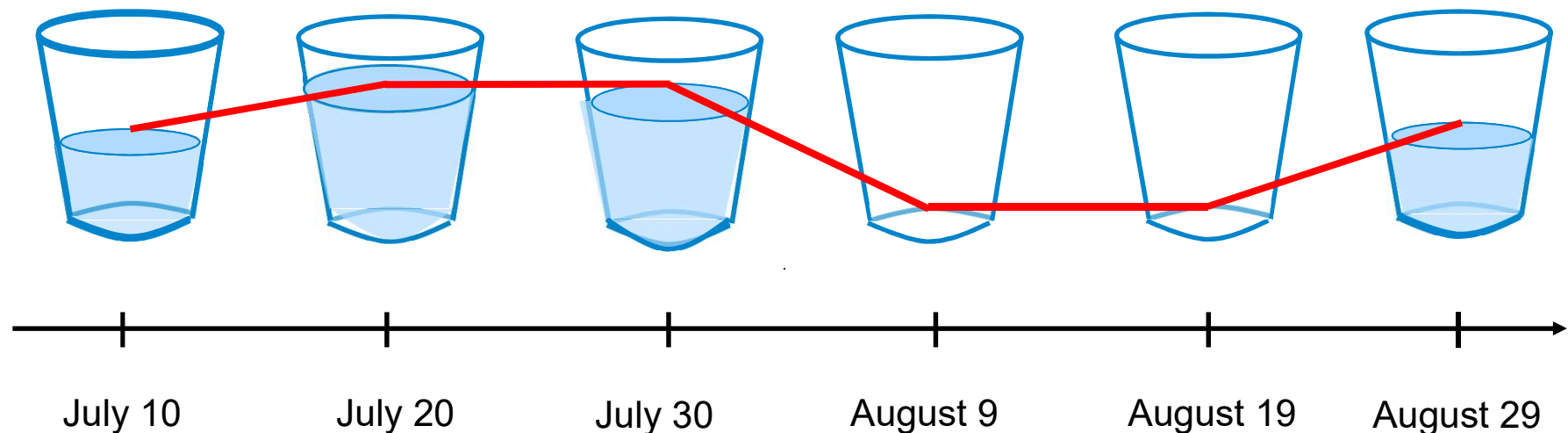
- Put a bucket in the back garden and record rainfall at regular intervals (e.g. every 10 days record rainfall and empty bucket)



## The Problem

### Approach 1

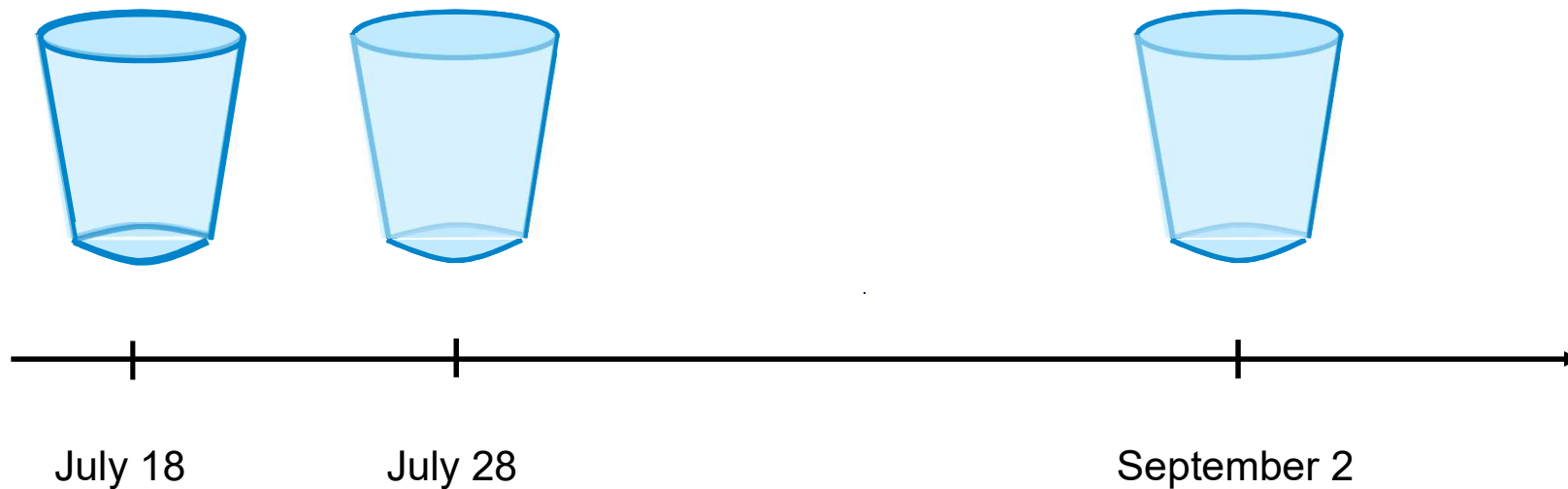
- **Advantage:** Easy estimation
- **Disadvantage:** inefficient (need to check and empty the bucket regularly even during the dry season and so...no summer vacations!)



## The Problem

### Approach 2

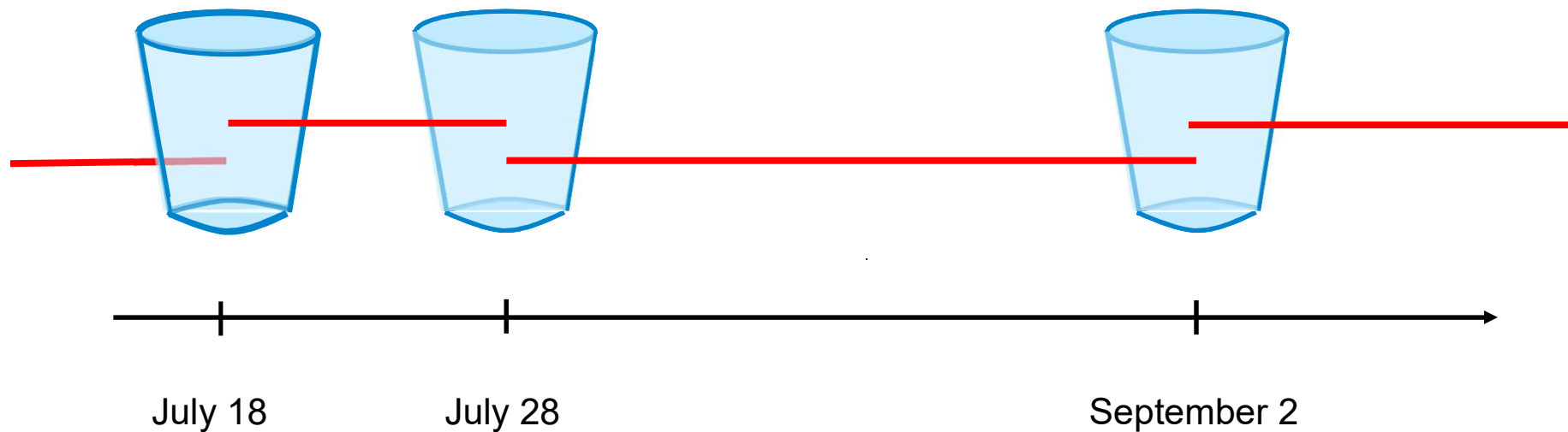
- Only record the day when the bucket is full and then empty it



## The Problem

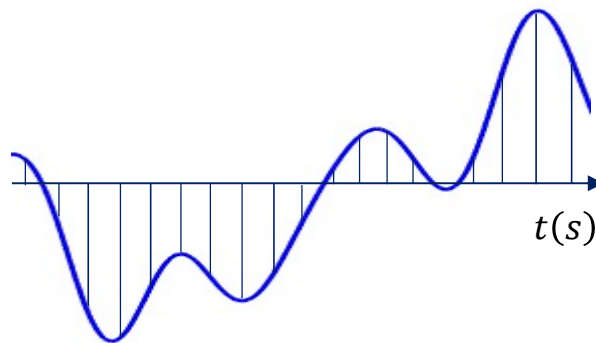
### Approach 2

- **Advantage:** Very efficient (we can go on holiday in August!)
- **Disadvantage:** estimation of rainfall over the period is more complicated



## Sampling

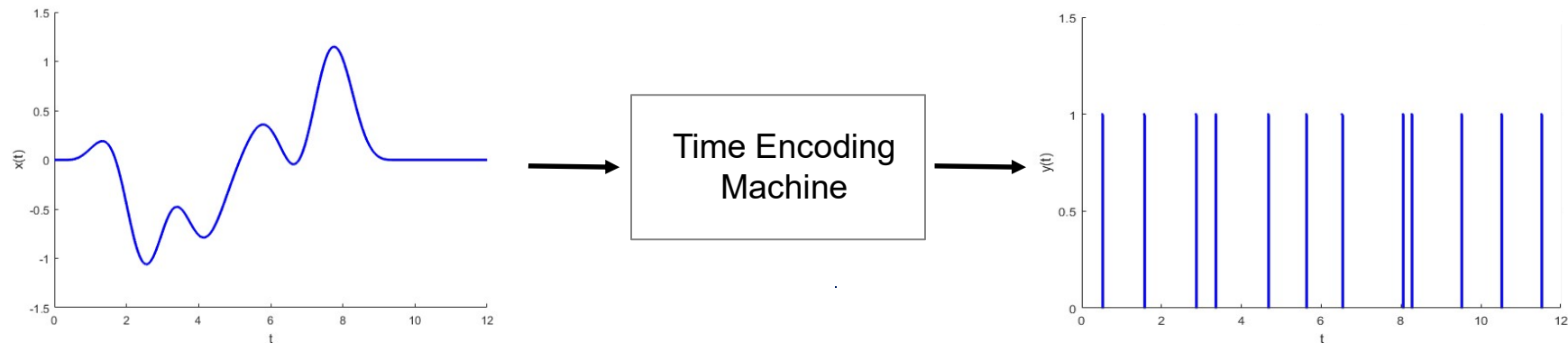
- These two approaches represent two very different ways to sample a continuous phenomenon
- [Approach 1](#) is what we engineers do and is equivalent to the traditional amplitude-based uniform sampling





# Sampling

- Approach 2 maps analogue information into a time sequence and is used by nature (e.g., **integrate-and-fire neurons**)



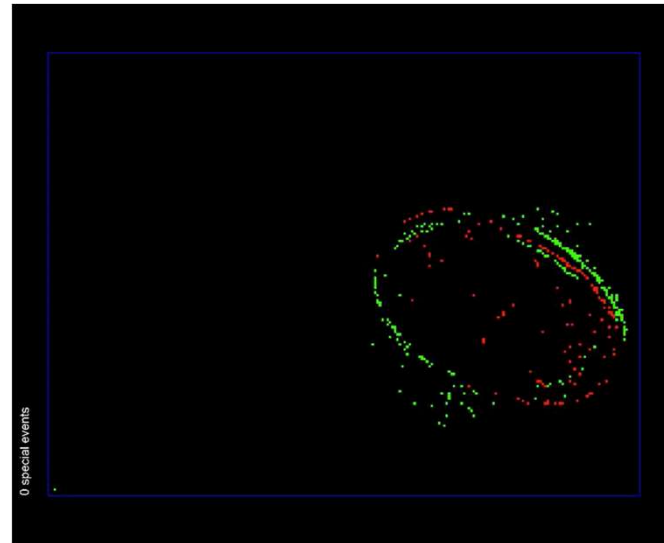
## Motivation

Time encoding appears in nature, as a mechanism used by neurons to represent sensory information as a sequence of action potentials, allowing them to process information **very efficiently**.



## Time-Based Sampling

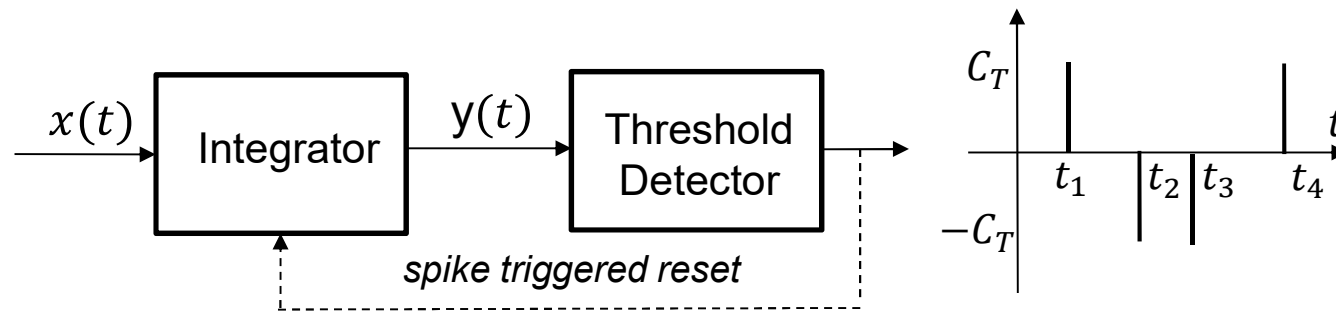
- Acquisition systems inspired by time-based sampling, such as [event-based vision sensors](#), are emerging in a variety of new scenarios (e.g. see Toby Delbruck web page)



Videos taken from Inivation.com (see also Toby Delbruck web page)

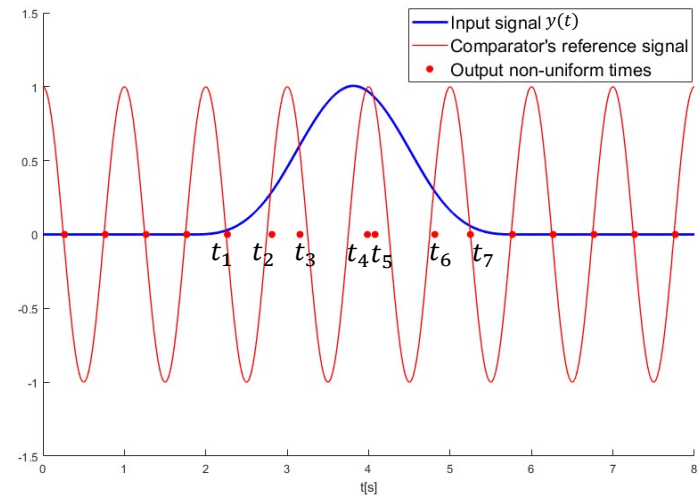
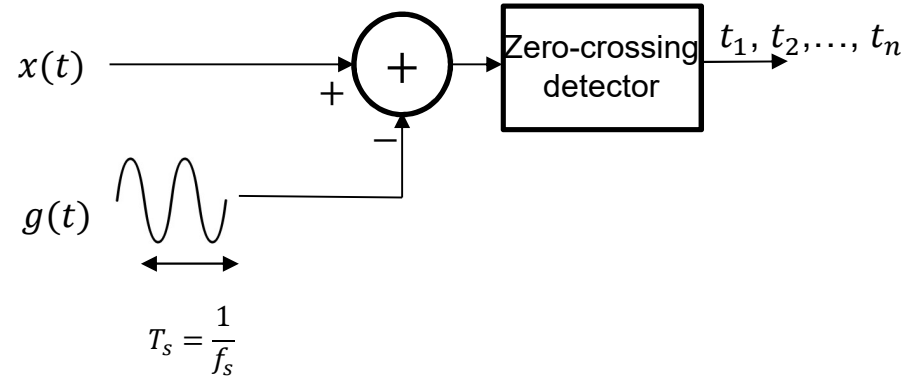
# Time-encoding machines

## Integrate-and-fire System



# Time-encoding machines

## Comparator System



- At the crossing times,  $x(t_n) - g(t_n) = 0$  hence  $x(t_n) = g(t_n)$ .

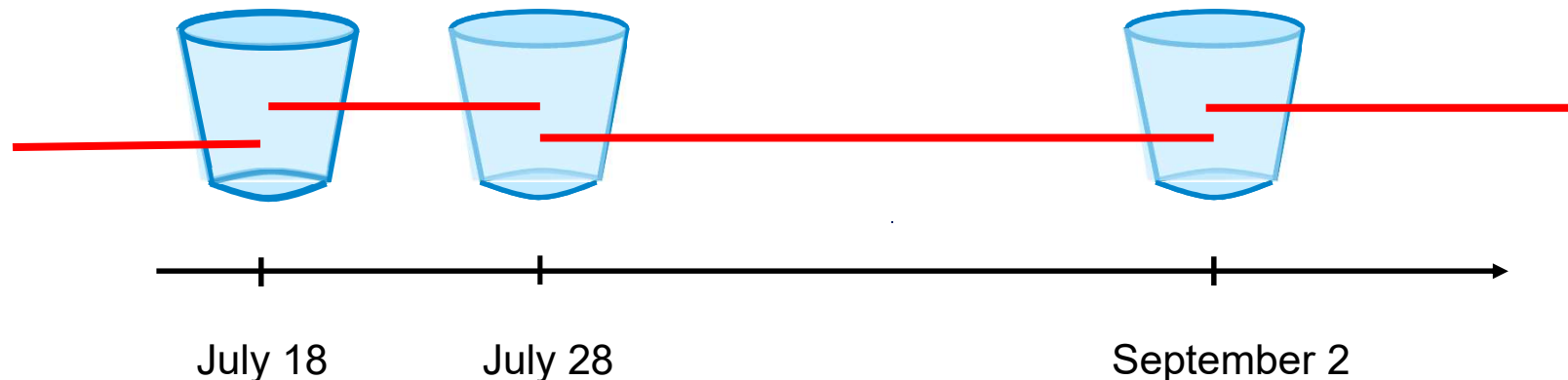
## Reconstruction from time-encoded information

- Given the retrieved non-uniform samples  $x(t_1), x(t_2), \dots, x(t_n)$  can we reconstruct  $x(t)$ ?
- This is a classical problem in non-uniform sampling
- Assume that  $x(t)$  belongs to a shift-invariant space (e.g.,  $x(t)$  is bandlimited,  $x(t) = \sum_k c_k \varphi(t - k)$ ) then, if the density of samples  $D \geq 1$ , perfect reconstruction is possible<sup>1</sup>

<sup>1</sup>A. Aldroubi and K. Grochenig, “Non-Uniform Sampling and Reconstruction in shift-invariant spaces”  
SIAM Review 2001

## Reconstruction from time-encoded information

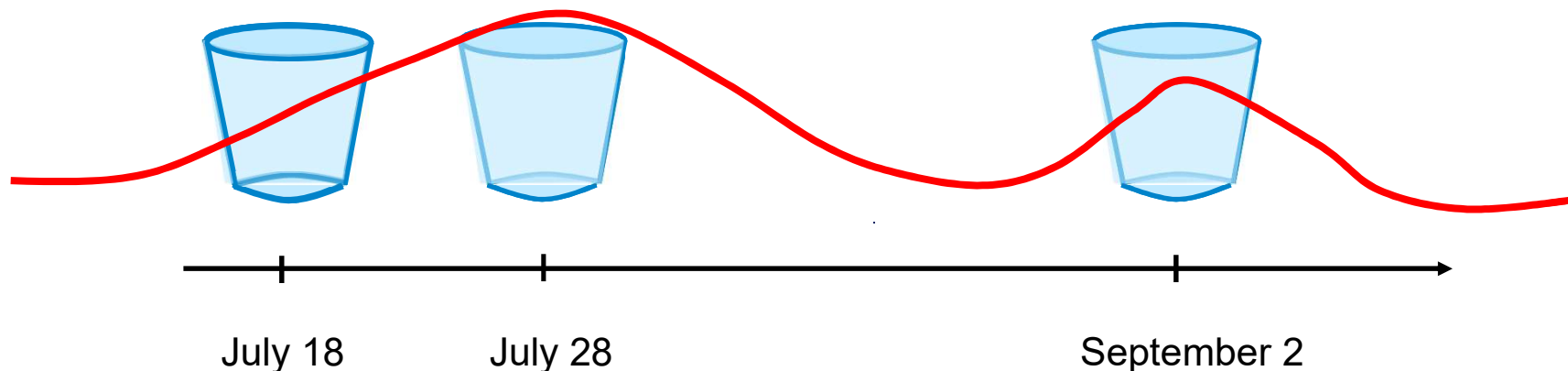
- **Key result:**<sup>1</sup> if the density of samples  $D \geq 1$  then perfect reconstruction can be using an iterative approach proposed by Aldroubi and Grochenig<sup>1</sup>



<sup>1</sup>A. Aldroubi and K. Grochenig, "Non-Uniform Sampling and Reconstruction in shift-invariant spaces" SIAM Review 2001

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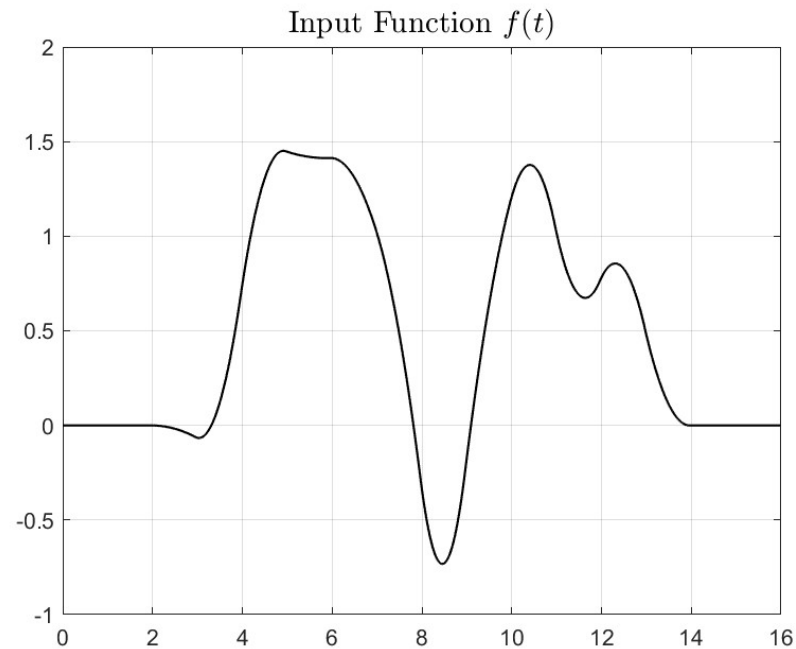


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## Reconstruction from time-encoded information

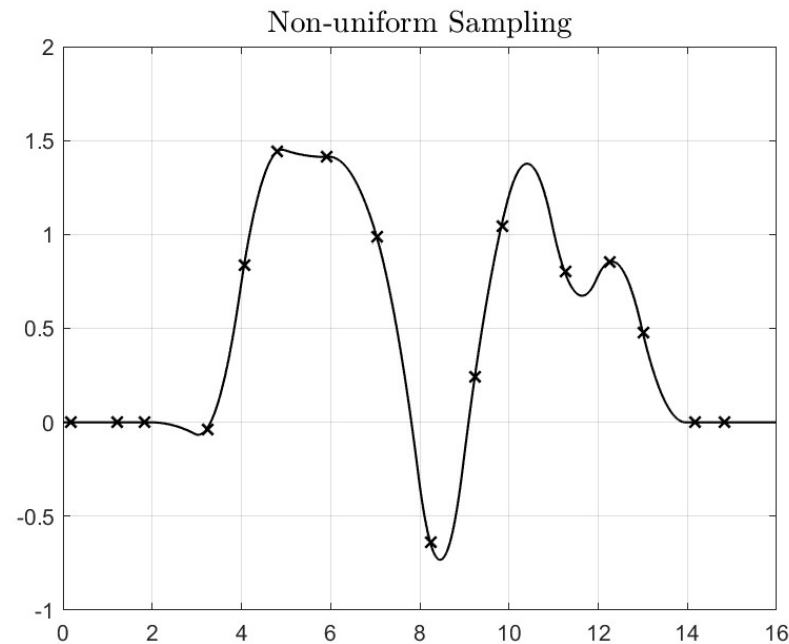
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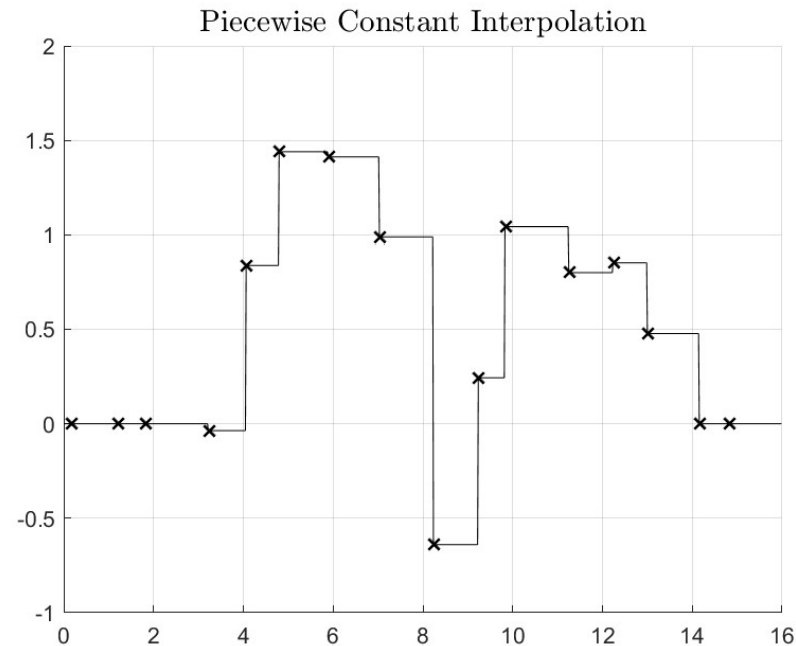
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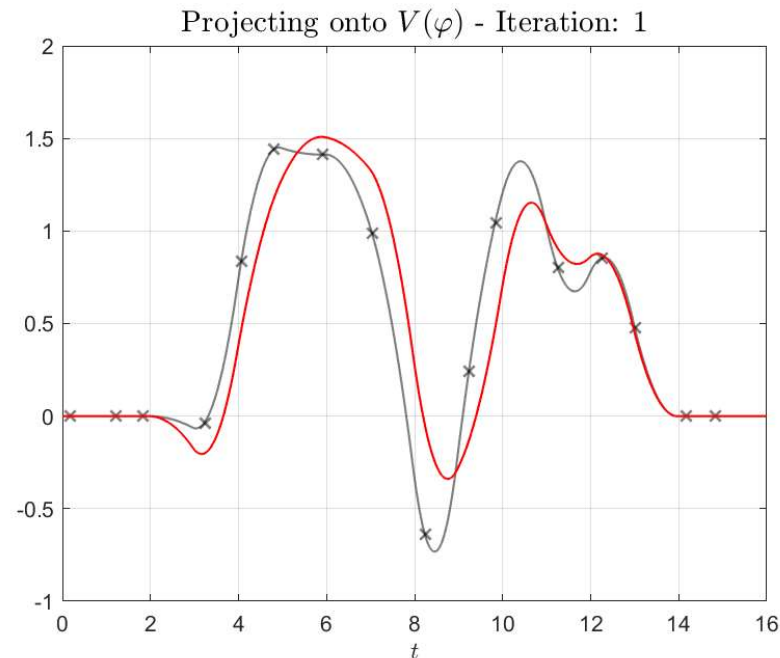
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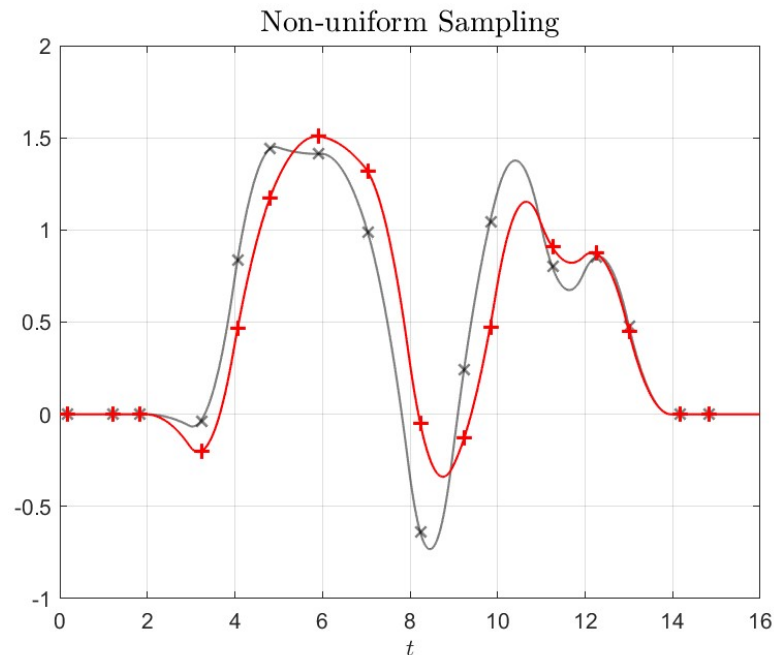
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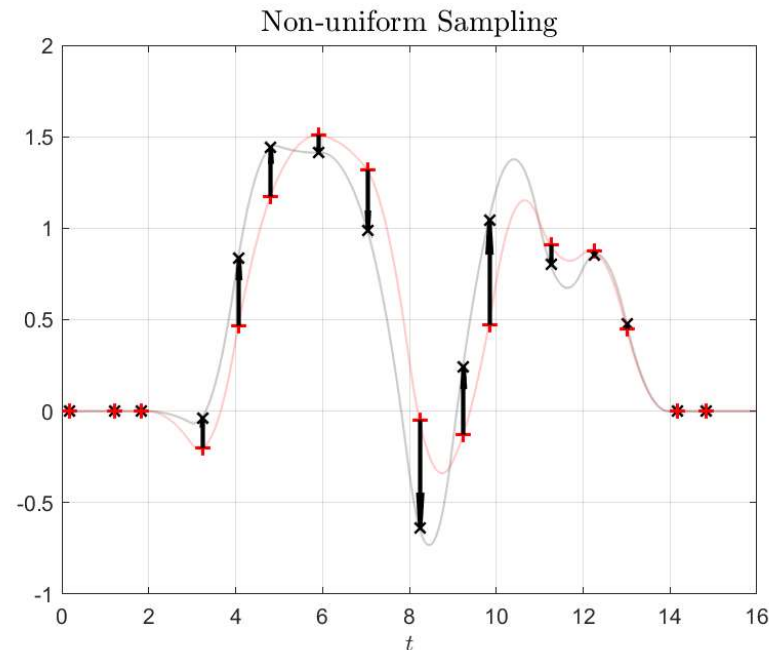
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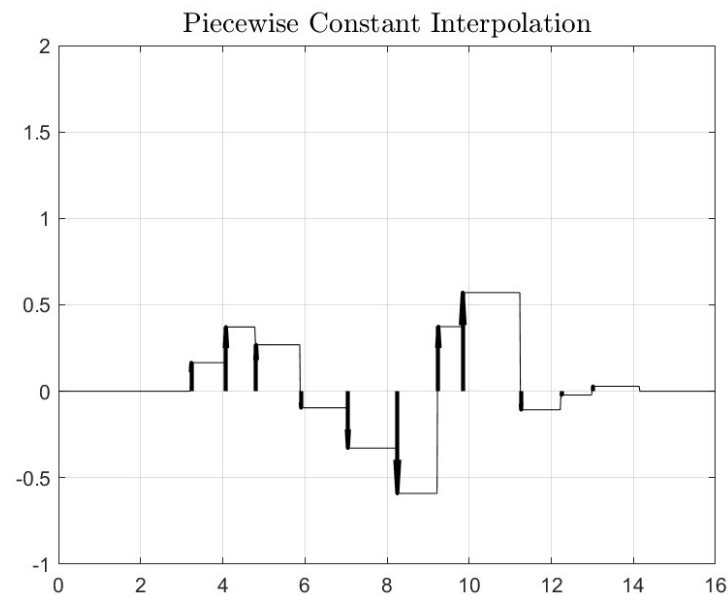
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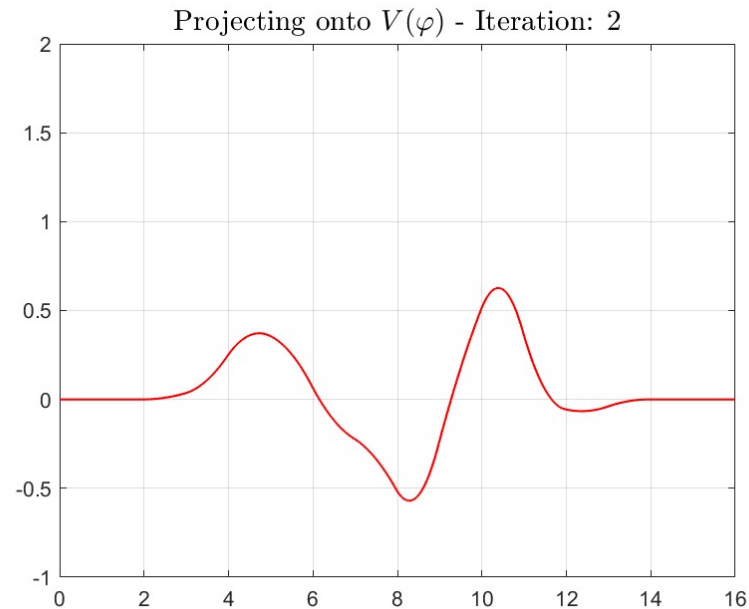
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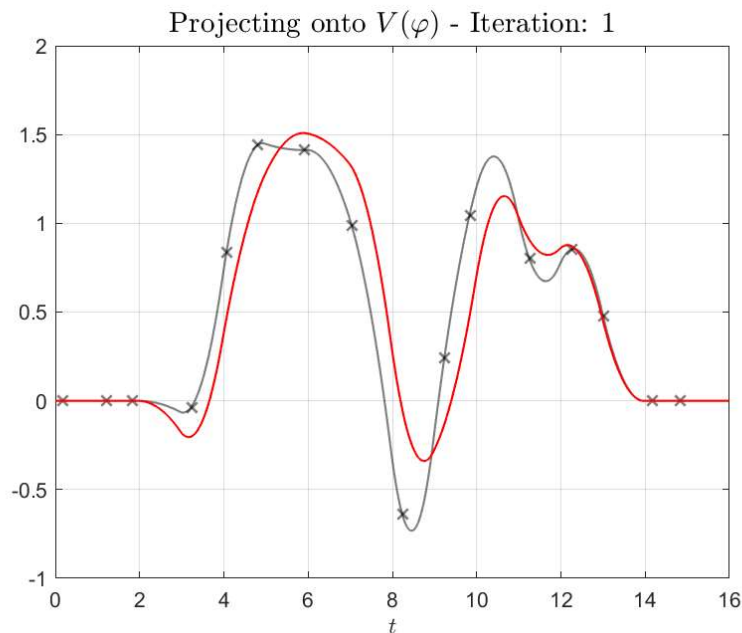


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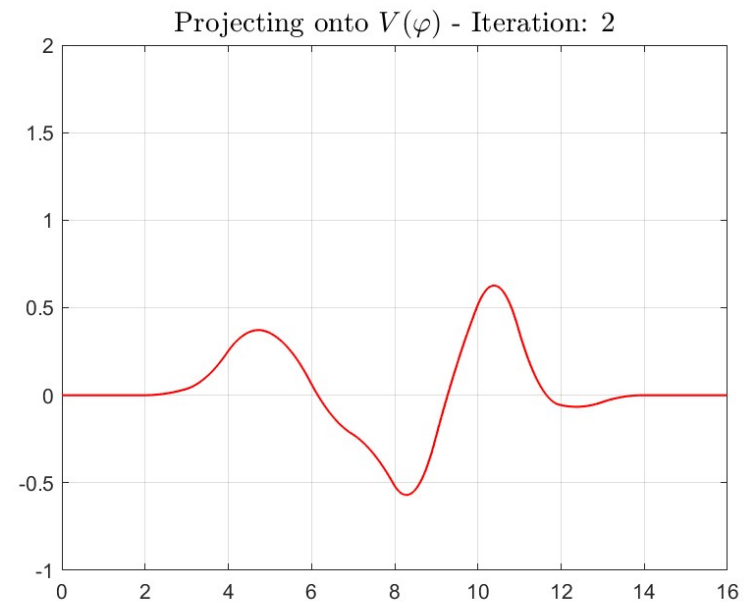


# Reconstruction from time-encoded information

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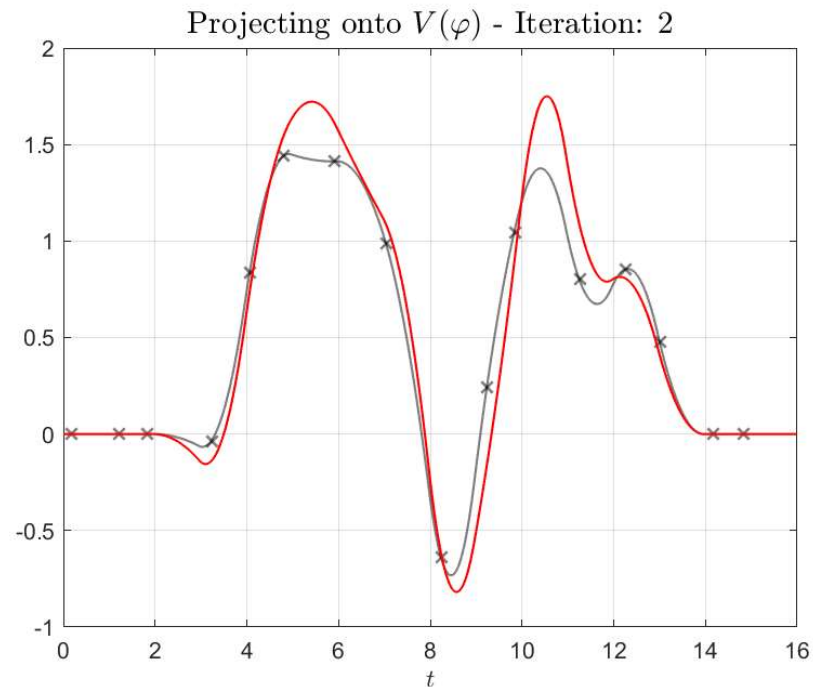
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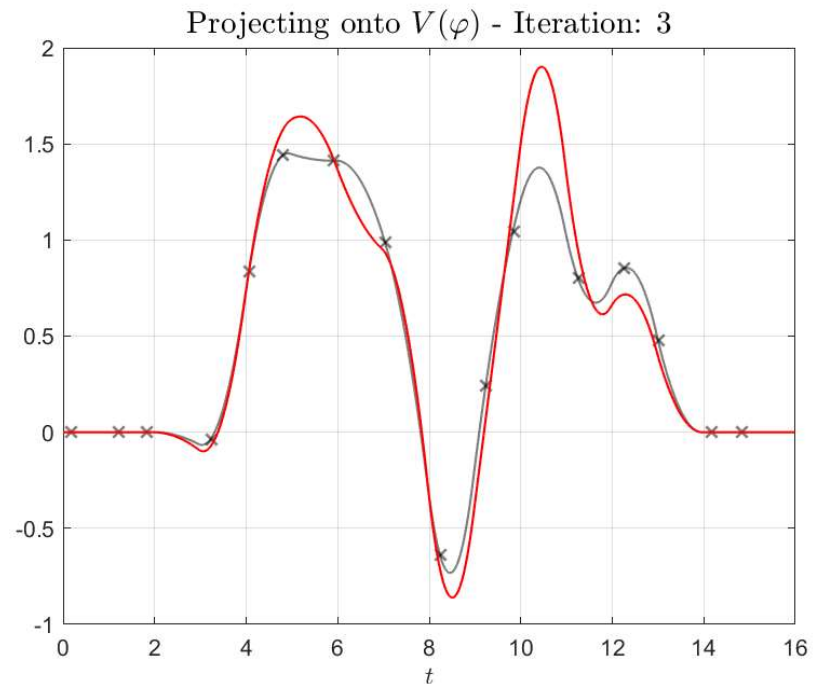
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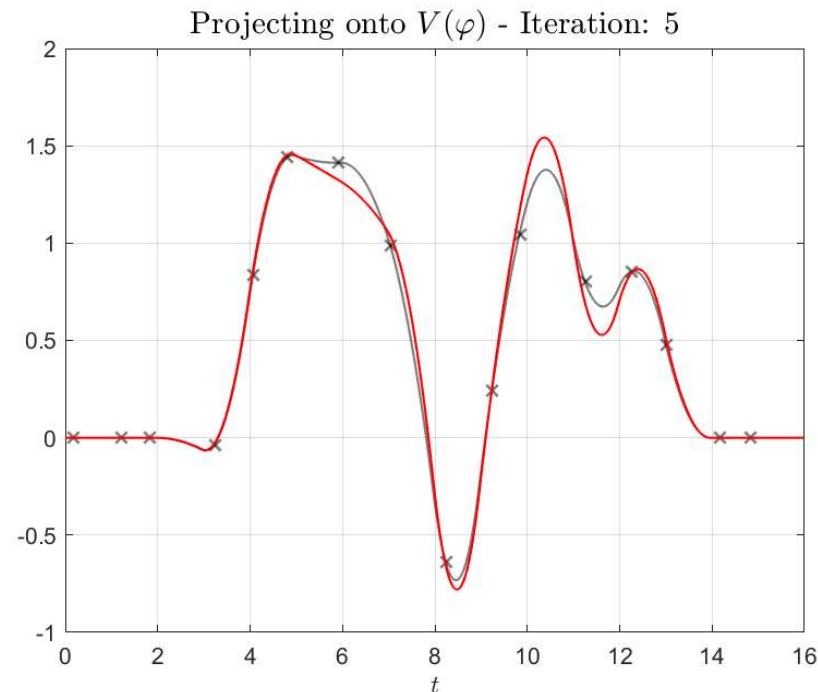
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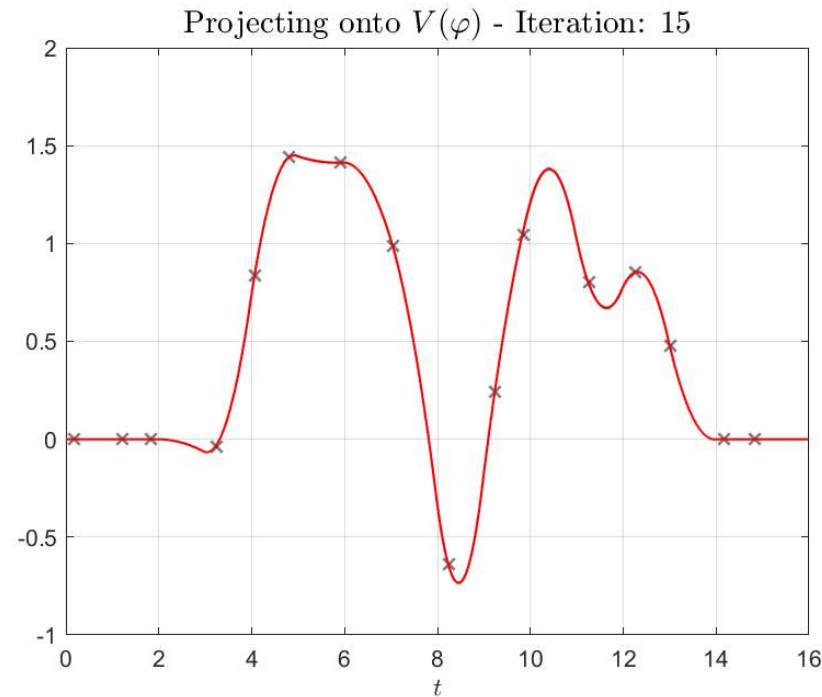
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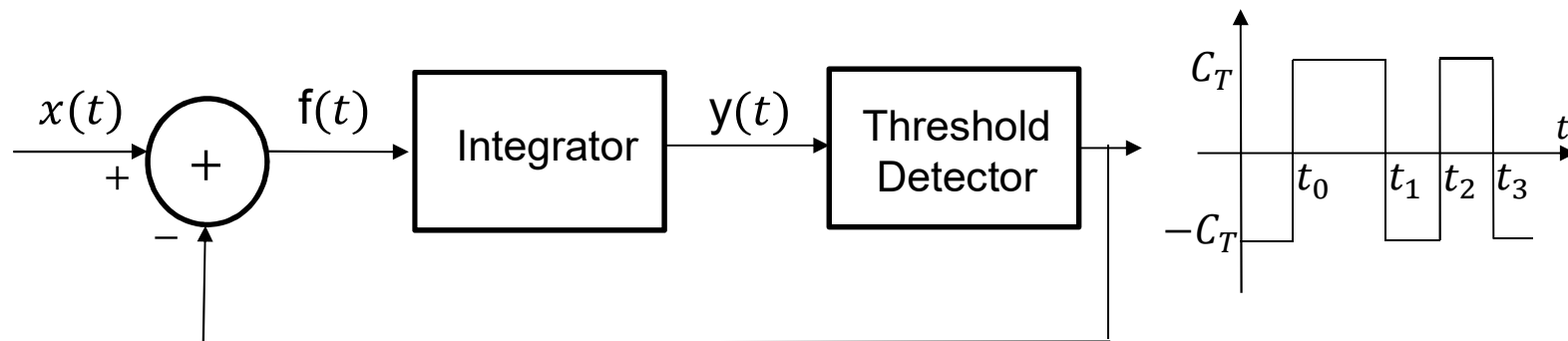
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## Reconstruction from time-encoded information

- **Key result:** if the density of samples  $D \geq 1$  then  $K_{t_j}(t)$  form a basis
- **Key Issue 1:** In the case of uniform sampling the density is  $D = 1$ . This means that current TEMs are **less** energy efficient than uniform sampling!
- **Key Issue 2:** Cannot sample sparse (non-bandlimited) signals with the current methods.

## Reconstruction from time-encoded information

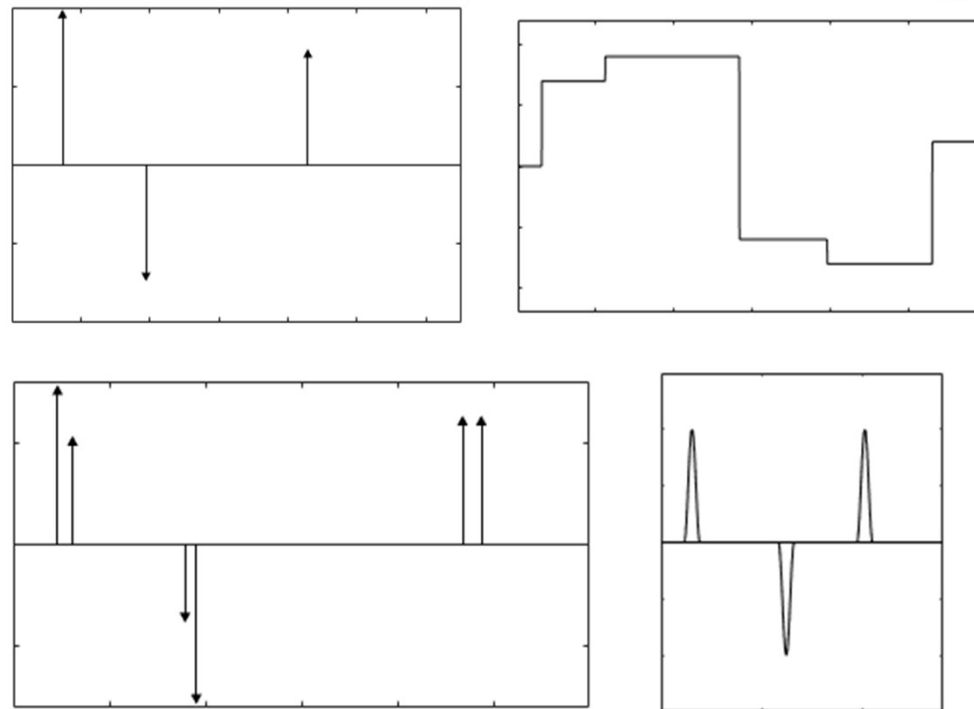
- For integrate-and-fire machines exact reconstruction proved here: A. A. Lazar and L. T. Toth, "Time encoding and perfect recovery of bandlimited signals", ICASSP 2003



See also: Gauntier-Vetterli-2014, Adam et al 2019,

## Sparse Sampling - Signals

- We consider sparse parametric signals (i.e., signals with finite rate of innovation<sup>2</sup>).
- Key issue is how to retrieve the free parameters of these signals for time-based information

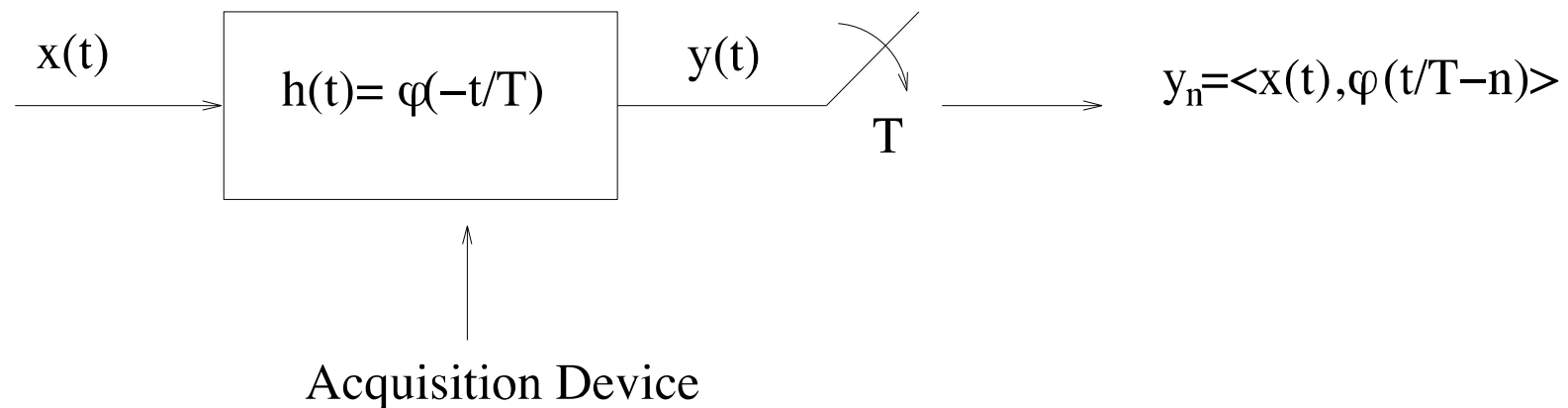


<sup>2</sup>Vetterli Marziliano Blu, Sampling Signals with Finite Rate of Innovation, IEEE Trans. on Signal Processing, June 2002



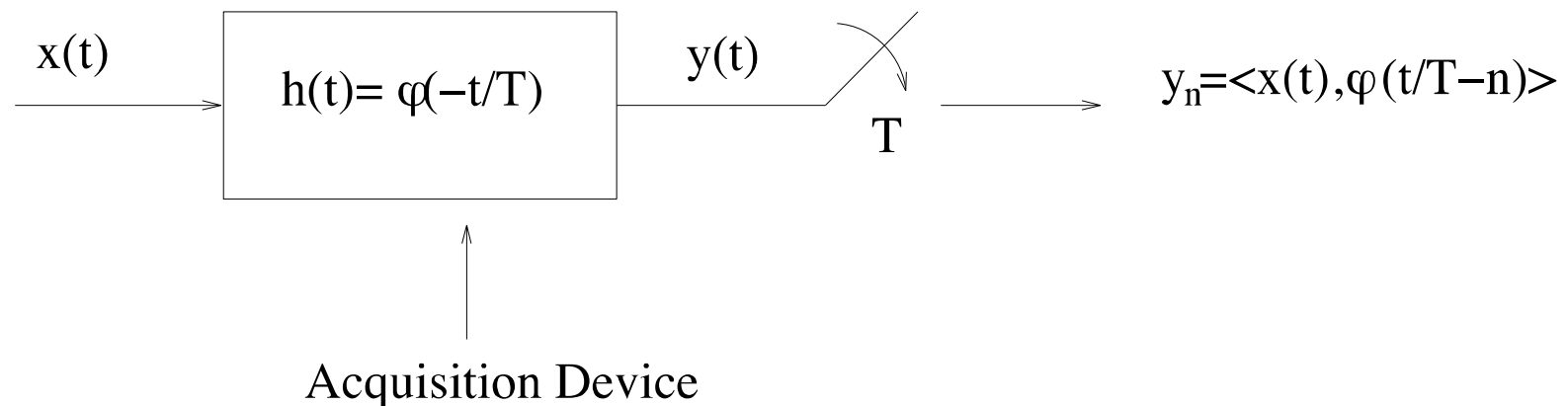
## Sparse Sampling - Acquisition

- In sparse sampling, the acquisition device is used to ‘spread the innovation’
- Reconstruction process is non-linear
- These two ingredients are necessary to time-encode sparse non-bandlimited signals

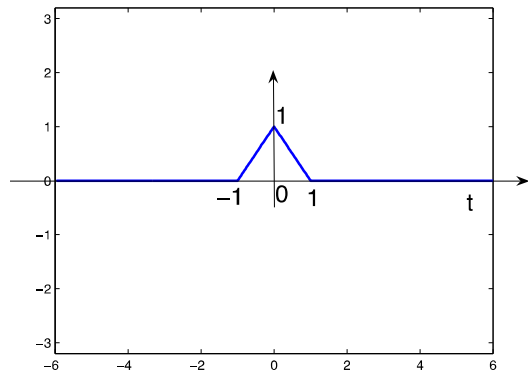


## Sparse Sampling

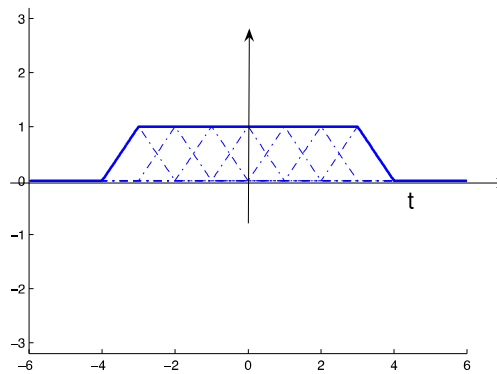
- We leverage two main ideas from sampling sparse signals with **finite rate of innovations**:
  - The sampling kernels can reproduce polynomials or exponentials
  - Reconstruction is achieved using Prony's method



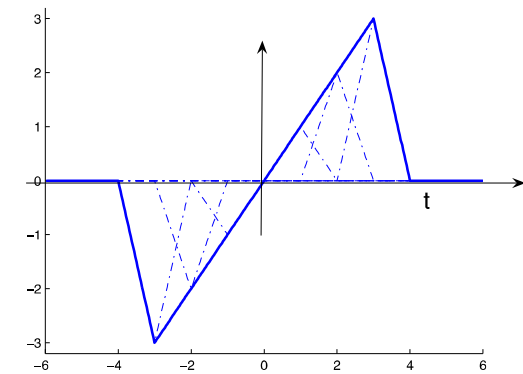
# Reproduction of Polynomials



$\beta_1(t)$



$c_{0,n} = (1, 1, 1, 1, 1, 1, 1)$

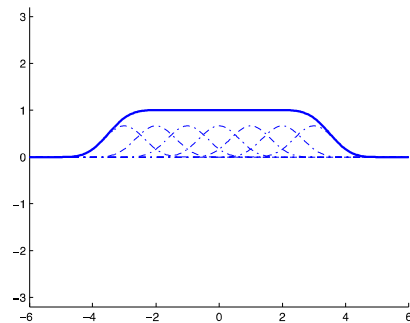


$c_{1,n} = (-3, -2, -1, 0, 1, 2, 3)$

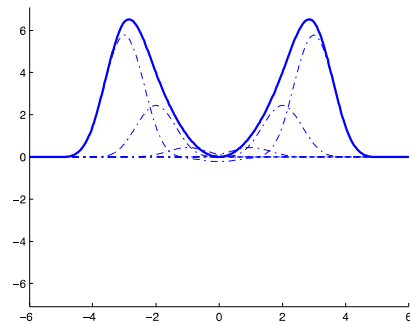
The linear spline reproduces polynomials up to degree  $L=1$ :  $\sum_n c_{m,n} \beta_1(t-n) = t^m \quad m = 0, 1$ , for a proper choice of coefficients  $c_{m,n}$  (in this example  $n = -3, -2, \dots, 1, 2, 3$ ).

**Notice:**  $c_{m,n} = \langle \tilde{\varphi}(t-n), t^m \rangle$  where  $\tilde{\varphi}(t)$  is biorthogonal to  $\varphi(t)$ :  $\langle \tilde{\varphi}(t), \varphi(t-n) \rangle = \delta_n$ .

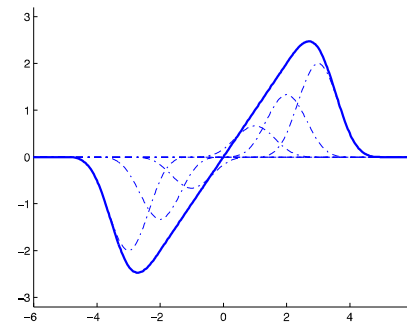
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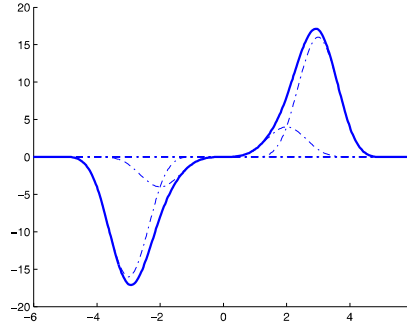
$$c_{0,n} = (1, 1, 1, 1, 1, 1, 1)$$



$$c_{2,n} \sim (8.7, 3.7, 0.7, -0.333, 0.7, 3.7, 8.7)$$



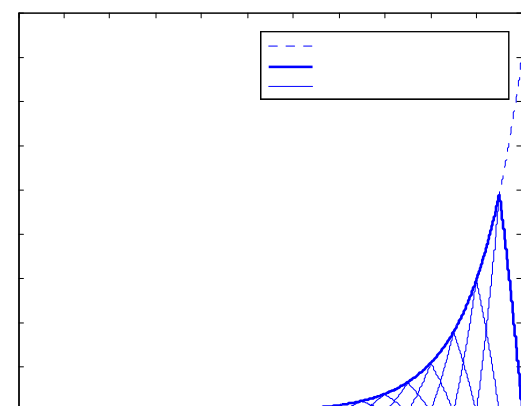
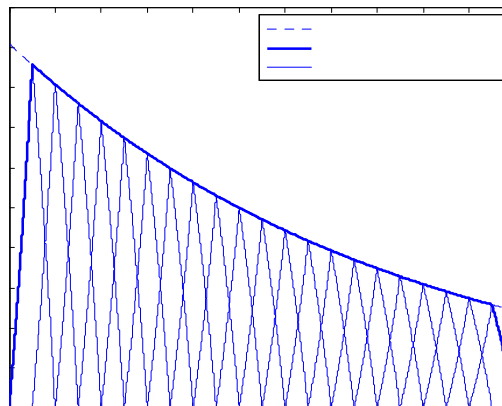
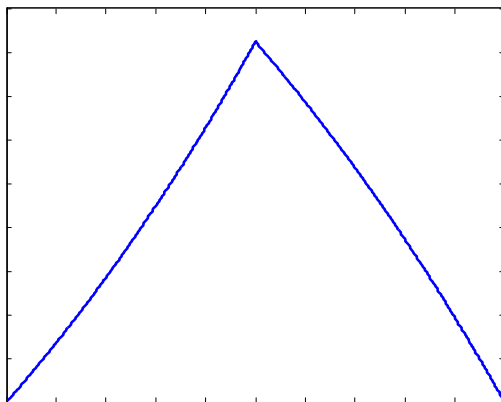
$$c_{1,n} = (-3, -2, -1, 0, 1, 2, 3)$$



$$c_{3,n} \sim (-24, -6, -0.001, 0, 0.001, 6, 24)$$

The cubic spline reproduces polynomials up to degree  $L=3$ :  $\sum_n c_{m,n} \beta_3(t-n) = t^m \quad m = 0, 1, 2, 3$ .

## Reproduction of Exponentials



Here the E-spline is of second order and reproduces the exponential  $e^{\alpha_0 t}$ ,  $e^{\alpha_1 t}$ : with  $\alpha_0 = -0.06$  and  $\alpha_1 = 0.5$ .

## From Samples to Signals

- ▶ Compute a linear combination of the samples:  $s_m = \sum_n c_{m,n} y_n$  for some choice of coefficients  $c_{m,n}$  that reproduce polynomials or exponentials
- ▶ Because of **linearity** of inner product, we have that

$$\begin{aligned} s_m &= \sum_n c_{m,n} y_n \\ &= \sum_n c_{m,n} \langle x(t), \varphi(t/T - n) \rangle \quad m = 0, 1, \dots, L. \\ &= \langle x(t), \sum_n c_{m,n} \varphi(t/T - n) \rangle \quad m = 0, 1, \dots, L. \end{aligned}$$

- ▶ Given the proper choice of coefficients, we have that  $\sum_n c_{m,n} \varphi(t/T - n) = e^{j\omega_m t/T}$

## From Samples to Signals

Then

$$\begin{aligned} s_m &= \sum_n c_{m,n} y_n \\ &= \langle x(t), \sum_n c_{m,n} \varphi(t/T - n) \rangle \\ &= \int_{-\infty}^{\infty} x(t) e^{j\omega_m t} dt, \quad m = 0, 1, \dots, L. \end{aligned}$$



## Sampling a stream of Diracs

- ▶ Assume  $x(t)$  is a stream of  $K$  Diracs on the interval of size  $N$ :  
 $x(t) = \sum_{k=0}^{K-1} x_k \delta(t - t_k)$ ,  $t_k \in [0, N)$ .
- ▶ We restrict  $j\omega_m = j\omega_0 + jm\lambda$   $m = 1, \dots, L$  and  $L \geq 2K$ .
- ▶ We have  $N$  samples:  $y_n = \langle x(t), \varphi(t - n) \rangle$ ,  $n = 0, 1, \dots, N - 1$ :
- ▶ We obtain

$$\begin{aligned} S_m &= \sum_{n=0}^{N-1} c_{m,n} y_n \\ &= \int_{-\infty}^{\infty} x(t) e^{j\omega_m t} dt, \\ &= \sum_{k=0}^{K-1} x_k e^{j\omega_m t_k} \\ &= \sum_{k=0}^{K-1} \hat{x}_k e^{j\lambda m t_k} = \sum_{k=0}^{K-1} \hat{x}_k u_k^m, \quad m = 1, \dots, L. \end{aligned}$$



## Prony's Method

- The quantity

$$s_m = \sum_{k=1}^K x_k e^{jm\omega_0\tau_k} = \sum_{k=1}^K x_k u_k^m \quad m = 1, \dots, L$$



is a sum of exponentials

- Retrieving the locations  $u_k$  and the amplitudes  $x_k$  from  $\{s_m\}_{m=1}^L$  is a classical problem in spectral estimation and was first solved by Gaspard de Prony in 1795.<sup>3</sup>
- Given the pairs  $\{x_k, u_k\}$  then  $\tau_k = (\ln u_k) / j\omega_0$ .

<sup>3</sup>P. Stoica and R. Moses. Spectral Analysis of Signals. 2005.

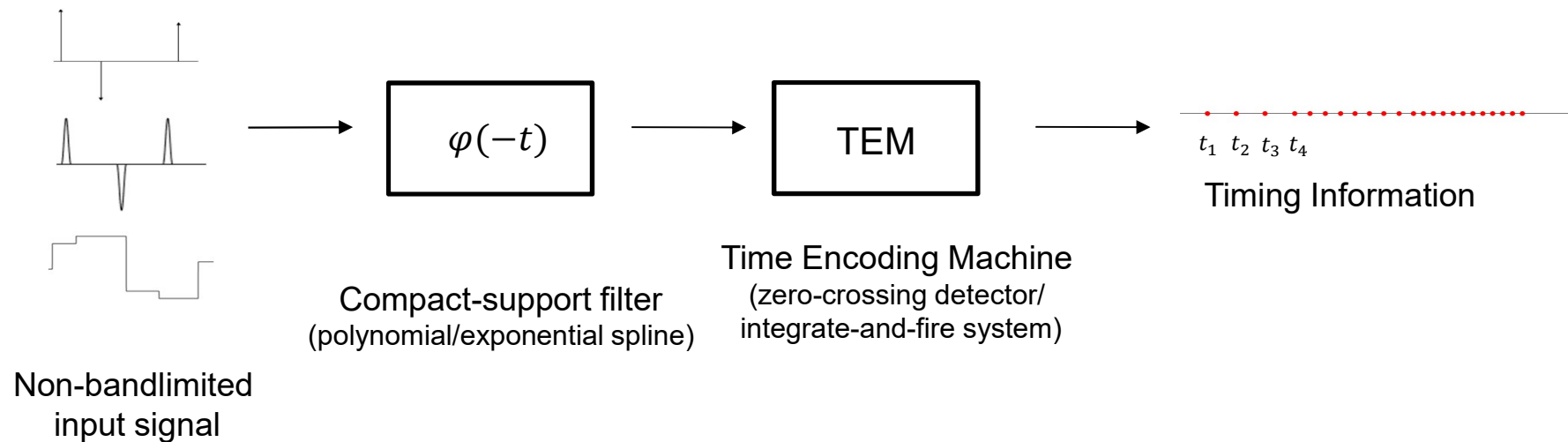
# Our approach for time decoding of signals

## Signals

- We consider sparse continuous-time signals like streams of diracs, stream of pulses or piecewise constant signals

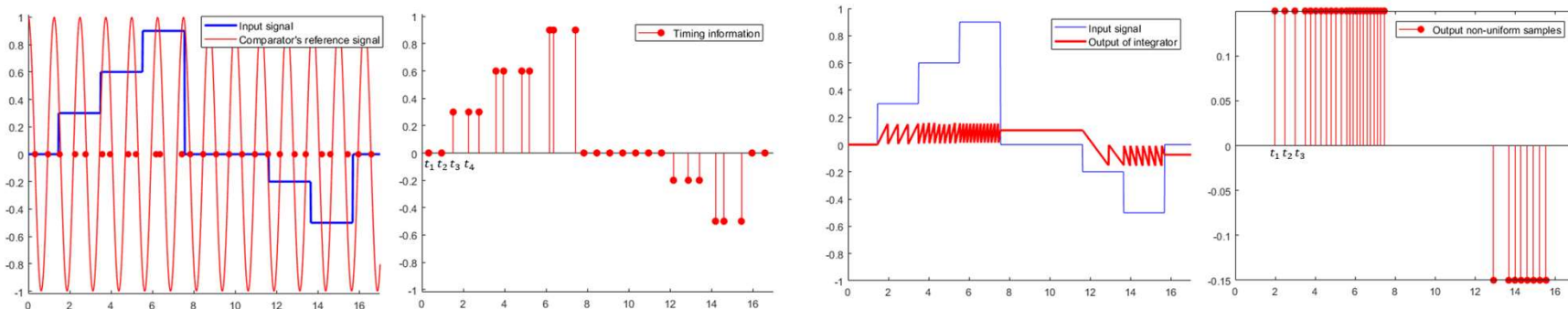
## Sensing Systems

- We filter before using a TEM



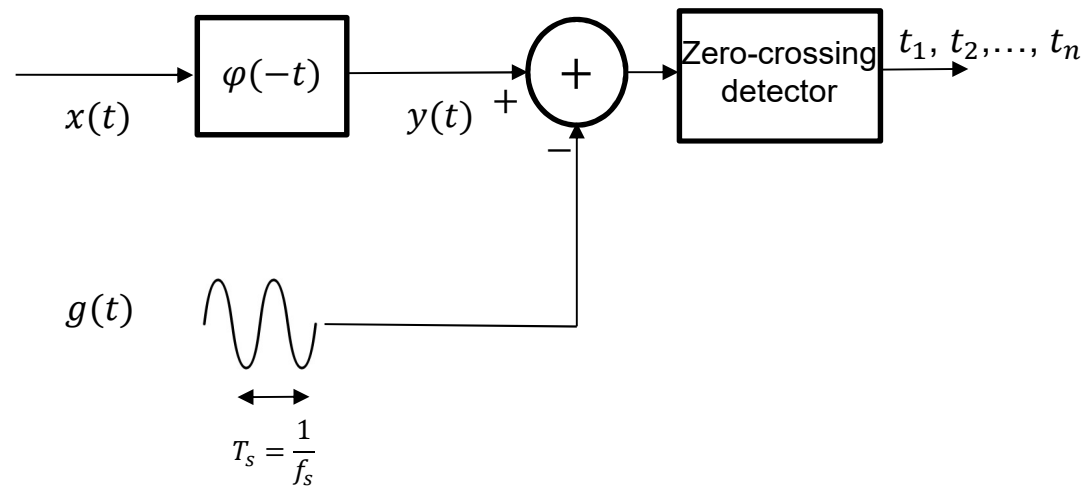
## Our approach for time decoding of signals

- Reconstruction of  $x(t)$  depends on the
  - sampling kernel  $\varphi(t)$
  - the density of time instants  $\{t_n\}$
- We achieve a sufficient density of output samples by imposing conditions on:
  - The frequency of the comparator's sinusoidal signal (**crossing TEM**).
  - The trigger mark of the integrator (**integrate-and-re TEM**).



## Our approach for time encoding of signals

### Comparator System



- At the crossing times,  $y(t_n) - g(t_n) = 0$  hence  $y(t_n) = g(t_n)$ .
- Moreover:

$$y(t_n) = \int x(\tau)\varphi(\tau - t_n) d\tau = \langle x(t), \varphi(t - t_n) \rangle$$

## Sampling Kernels (B-splines)

- The anti-causal version of the zero-order B-spline is defined as:

$$\beta_0(t) = \begin{cases} 1, & -1 \leq t \leq 0, \\ 0, & \text{otherwise.} \end{cases}$$

- The P-order B-spline can be computed as:

$$\beta_P(t) = \underbrace{\beta_0(t) * \beta_0(t) \dots * \beta_0(t)}_{P+1 \text{ times}},$$

- The P-order B-spline satisfies the Strang-Fix condition:

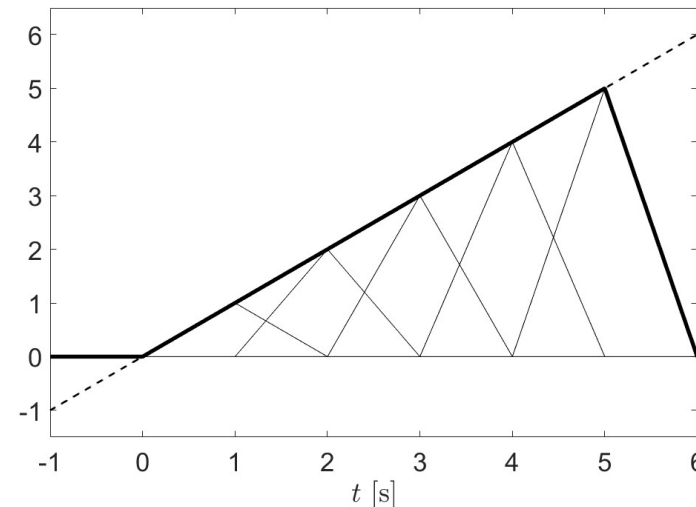
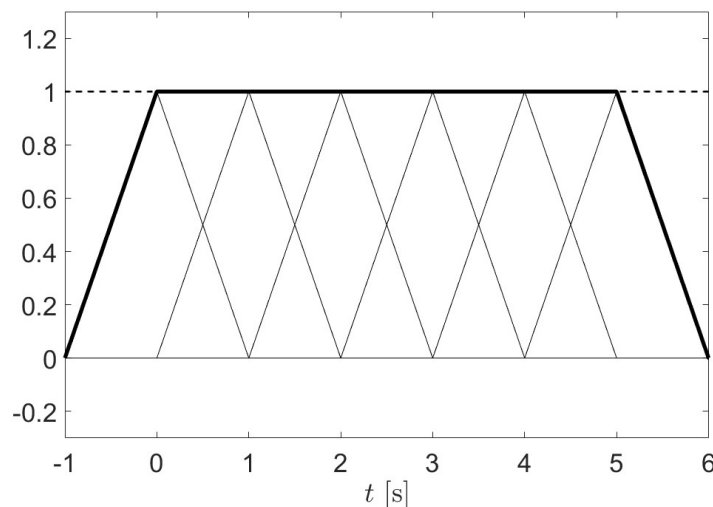
$$\sum_{n \in \mathbb{Z}} c_{m,n} \beta_P(t - n) = t^m,$$

where  $m \in \{0, 1, \dots, P\}$ , and for a proper choice of coefficients  $c_{m,n}$ .

## Sampling Kernels (B-splines)

### Polynomial Splines

- Linear combinations of uniform shifts of B-splines reproduce polynomial because the *'knots'* overlap and *'compensate'* each other.



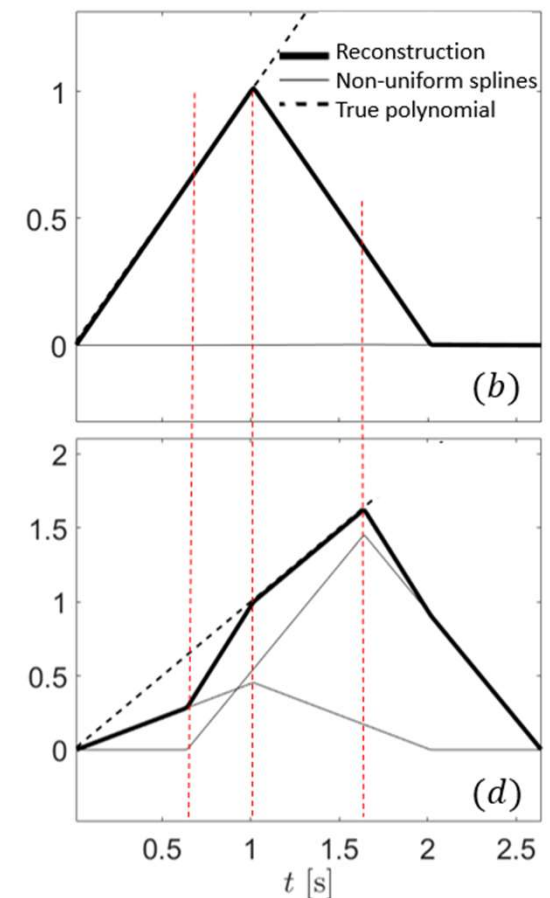
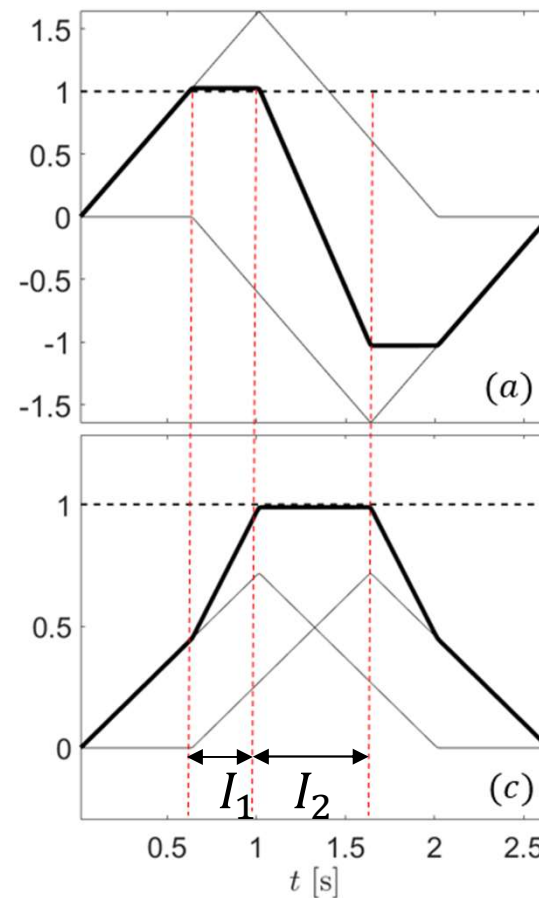
- Key insight:** in the case of non-uniform shifts, reproduction of polynomials is still possible locally in *'knot-free'* regions

## Sampling Kernels (B-splines)

**Key insight:** in the case of non-uniform shifts, reproduction of polynomials is still possible locally in 'knot-free' regions

Sketch of the argument:

- Each 'knot-free' piece of a spline of order  $d$  is a polynomial of degree  $d$
- $d$  overlapping splines can reproduce polynomial of maximum degree  $d$  in a 'knot-free' region



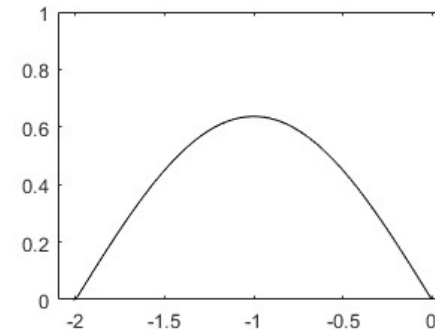
## Sampling Kernels (E-splines)

- Exponential Splines (E-splines) can reproduce exponentials:

$$\sum_{n \in \mathbb{Z}} c_{m,n} \varphi(t - n) = e^{-\alpha_m t}$$

- The first-order E-spline of support L is defined as:

$$\varphi(t) = \begin{cases} \frac{e^{\alpha_1 - \alpha_0}}{\alpha_1 - \alpha_0} e^{-\alpha_0 t} + \frac{e^{-\alpha_1 + \alpha_0}}{\alpha_0 - \alpha_1} e^{-\alpha_1 t}, & -L \leq t \leq \frac{-L}{2} \\ \frac{1}{\alpha_0 - \alpha_1} e^{-\alpha_0 t} + \frac{1}{\alpha_1 - \alpha_0} e^{-\alpha_1 t}, & \frac{-L}{2} \leq t \leq 0 \\ 0, & \text{otherwise} \end{cases}$$



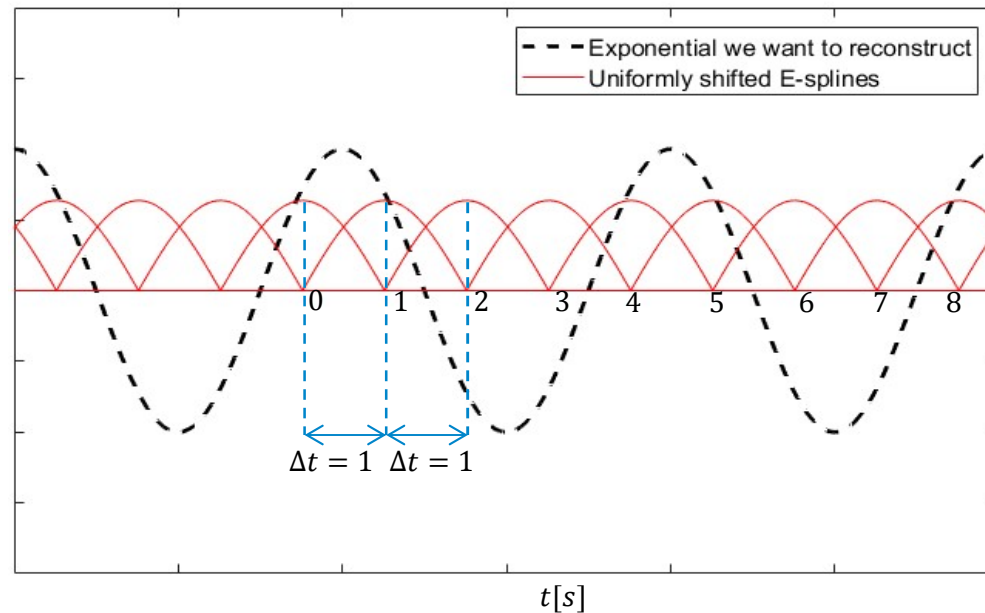
- This function can reproduce exponentials  $e^{-\alpha_0 t}$  and  $e^{-\alpha_1 t}$ .



## Sampling Kernels

- Reproduction of exponentials using uniform shifts of the first-order E-spline:

$$\sum_{n \in \mathbb{Z}} c_{m,n} \varphi(t - n) = e^{-\alpha_m t}$$

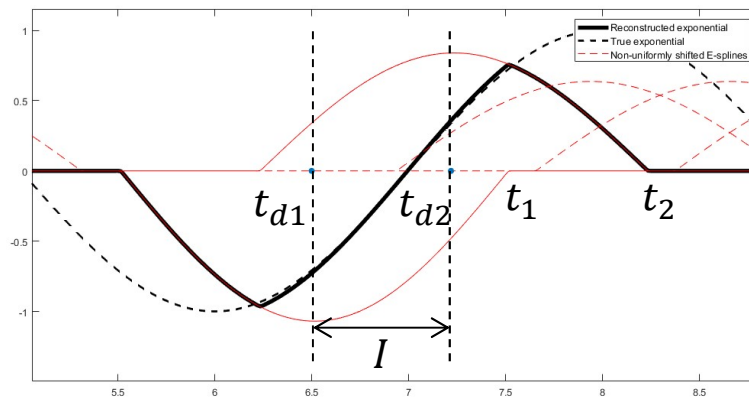


# Sampling Kernels

- Reproduction of exponentials can be achieved locally in  $I$ , using at least two non-uniform shifts of the E-spline:

$$\sum_{n=1}^N c_{m,n} \varphi(t - t_n) = e^{-\alpha_m t}, N \geq 2$$

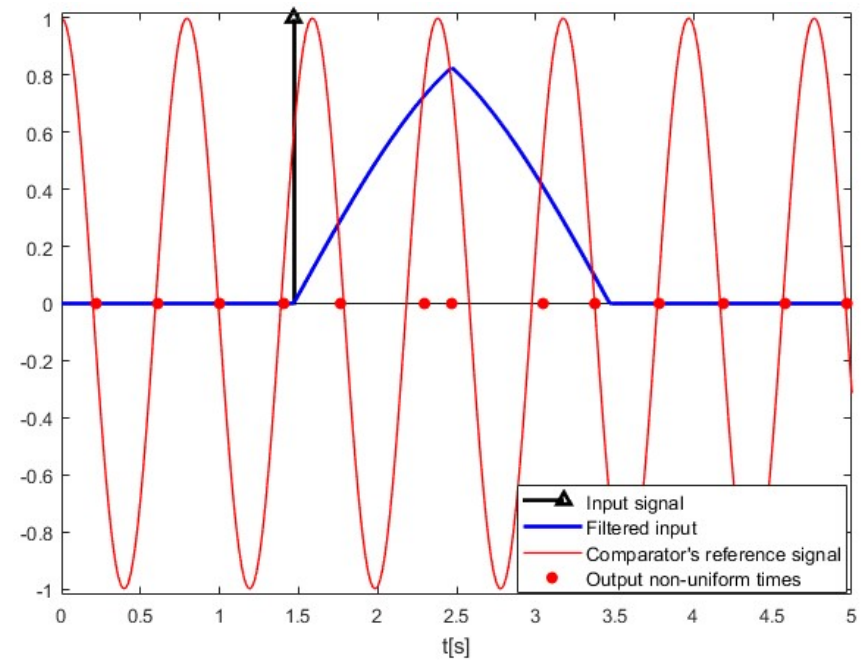
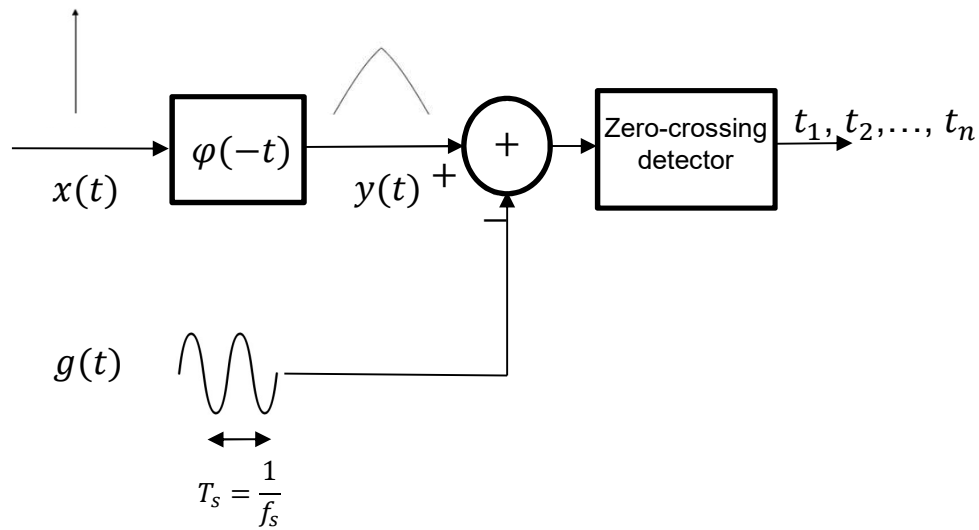
- The kernels should be continuous within that local interval  $I$ .



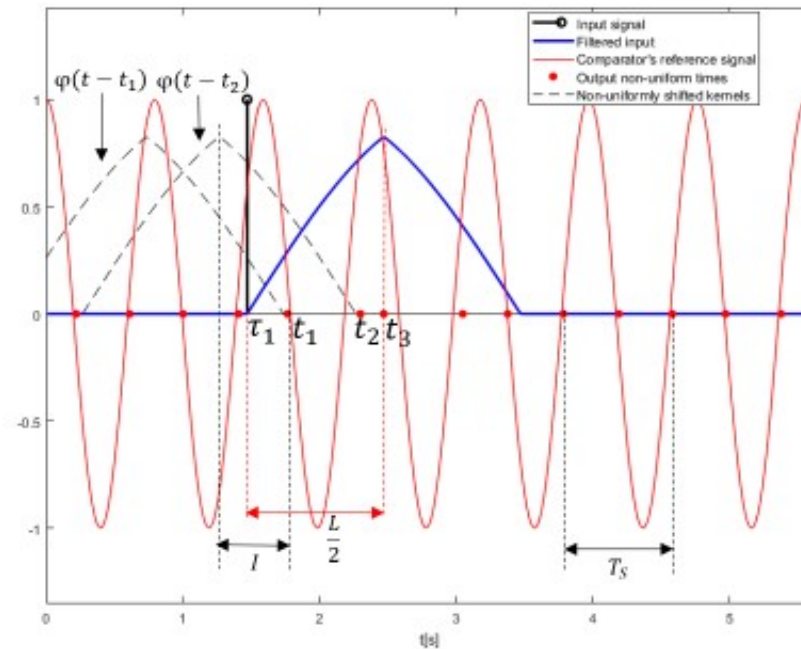
$t_{d1}$  - discontinuity of  $\varphi(t - t_1)$

$t_{d2}$  - discontinuity of  $\varphi(t - t_2)$

# Comparator System – One Input Spike



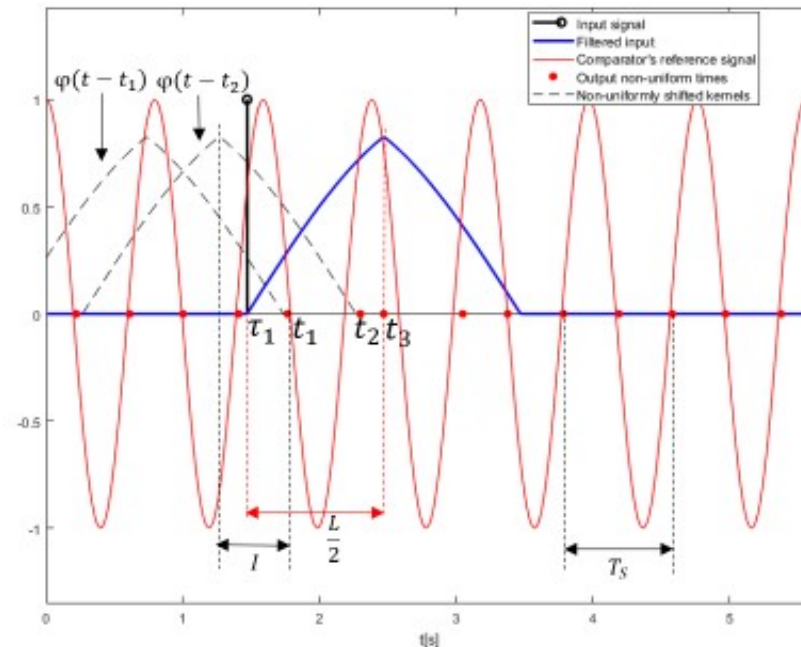
# Comparator System – One Input Spike



$$y(t_1) = \langle x(t), \varphi(t - t_1) \rangle$$

$$y(t_2) = \langle x(t), \varphi(t - t_2) \rangle$$

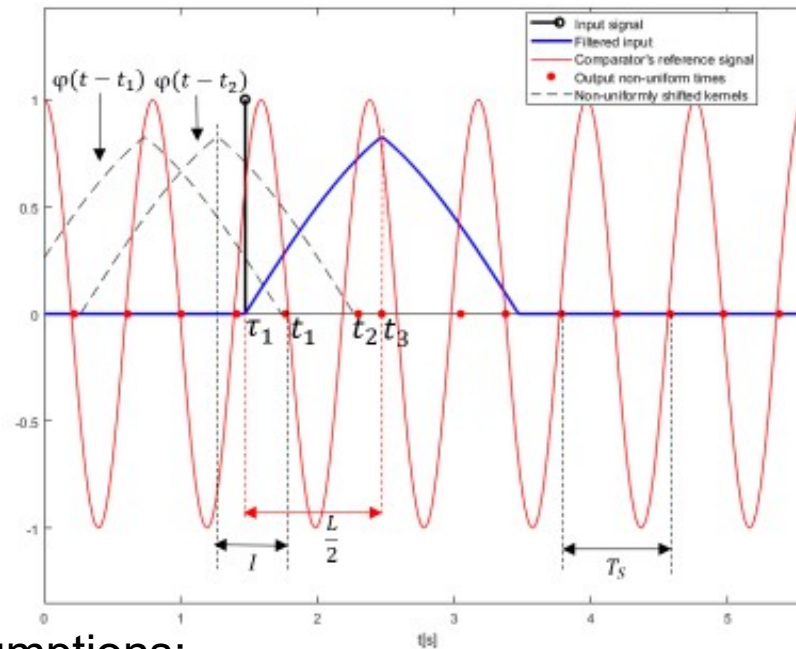
# Comparator System – One Input Spike



We assume:

- Amplitude of the Dirac  $|x_1| < 1$
- The sampling kernel  $\varphi(t)$  and its non-uniform shifts reproduce  $e^{j\omega_0 t}$  and  $e^{-j\omega_0 t}$  and  $0 < \omega_0 \leq \frac{\pi}{L}$  where  $L$  is the support of  $\varphi(t)$ .
- The frequency of the sinusoidal signal satisfies  $f_s > \frac{5}{2L}$ .

## Comparator System – One Input Spike



Under these assumptions:

- The first two timing locations satisfy  $t_1, t_2 \in [\tau_1, \tau_1 + \frac{L}{2}]$ . This means that  $\tau_1 \in [t_2 - \frac{L}{2}, t_1]$
- This is useful since in the interval  $I = [t_2 - \frac{L}{2}, t_1]$ , the shifted kernels  $\varphi(t - t_1)$  and  $\varphi(t - t_2)$  have no knots (so can reproduce exponentials or polynomials)

## Comparator System – One Input Spike

- We know we can find coefficients  $c_{m,n}^I$  such that:

$$\sum_{n=1}^2 c_{m,n}^I \varphi(t - t_n) = e^{j\omega_m t}, \text{ for } t \in I, m = 0, 1.$$

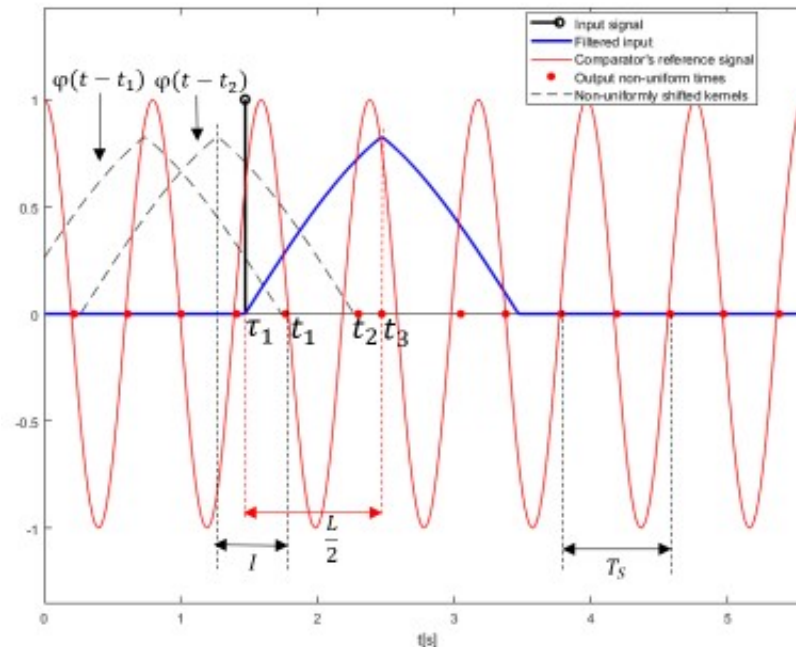
- We then have:

$$s_0 = \sum_{n=1}^2 c_{0,n}^I y(t_n) = \sum_{n=1}^2 c_{0,n}^I \langle x(t), \varphi(t - t_n) \rangle = \int_{-\infty}^{\infty} x(t) \sum_{n=1}^2 c_{0,n}^I \varphi(t - t_n) dt = x_1 e^{j\omega_0 \tau_1},$$

$$s_1 = \sum_{n=1}^2 c_{1,n}^I y(t_n) = \sum_{n=1}^2 c_{1,n}^I \langle x(t), \varphi(t - t_n) \rangle = \int_{-\infty}^{\infty} x(t) \sum_{n=1}^2 c_{1,n}^I \varphi(t - t_n) dt = x_1 e^{j\omega_1 \tau_1},$$

- Then  $\tau_1 = \frac{1}{j(\omega_1 - \omega_0)} \ln \frac{s_1}{s_2}$  and  $x_1 = s_0 / e^{j\omega_0 \tau_1}$

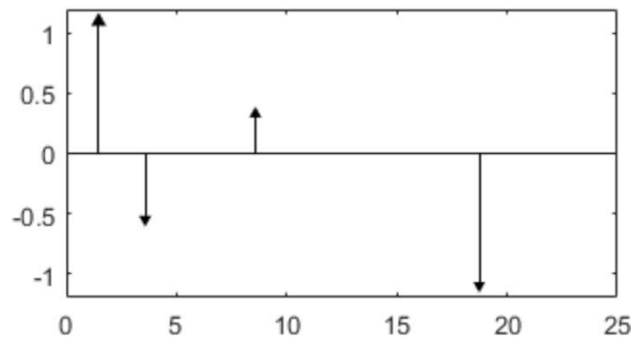
## Comparator System – One Input Spike



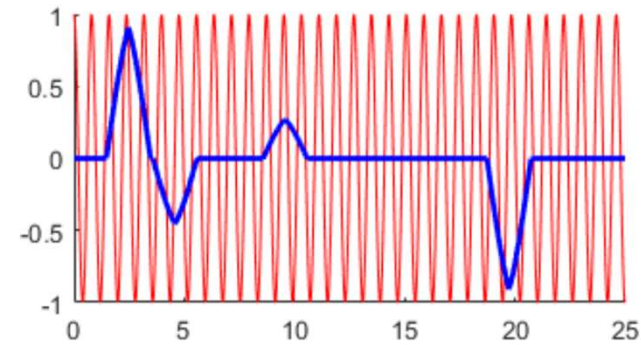
- We use Bolzano's intermediate value theorem to show that  $t_1, t_2 \in [\tau_1, \tau_1 + \frac{L}{2}]$
- Denote with  $h(t) = g(t) - y(t)$ , assume  $g(\tau_1) > 0$  and  $x_1 > 0$  then  $h(\tau_1) > 0$  and  $h(\tau_1 + \frac{T_s}{2}) = g(\tau_1 + \frac{T_s}{2}) - y(\tau_1 + \frac{T_s}{2}) < 0$ , this implies  $h(t_1) = 0$  for some  $t_1 \in [\tau_1, \tau_1 + \frac{T_s}{2}]$
- Similarly  $t_2 \in [\tau_1 + \frac{T_s}{2}, \tau_1 + \frac{5T_s}{4}]$
- Since  $T_s = \frac{1}{f_s} < \frac{2L}{5}$  then  $t_1, t_2 \in [\tau_1, \tau_1 + \frac{L}{2}]$



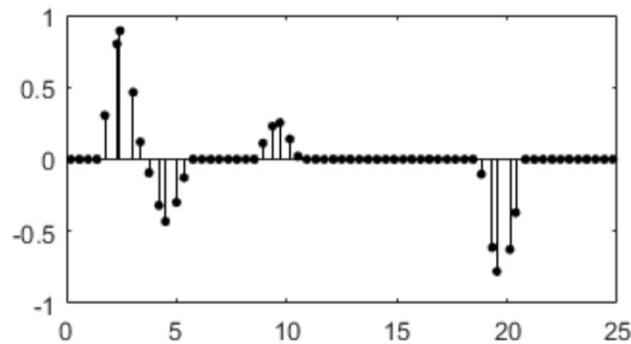
# Comparator System – Example



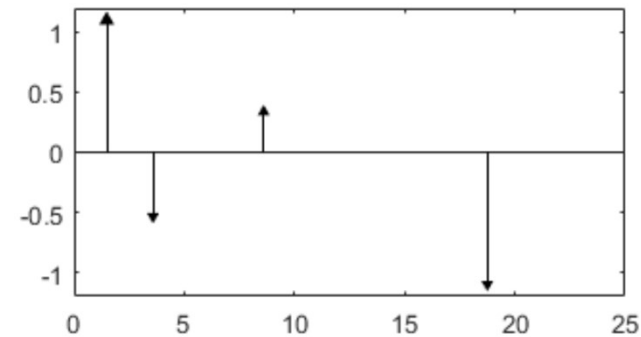
(a)



(b)



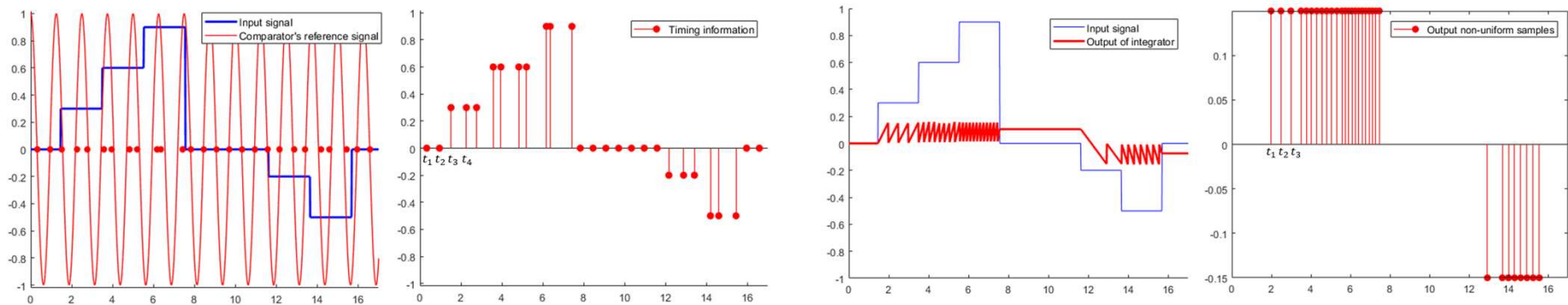
(c)



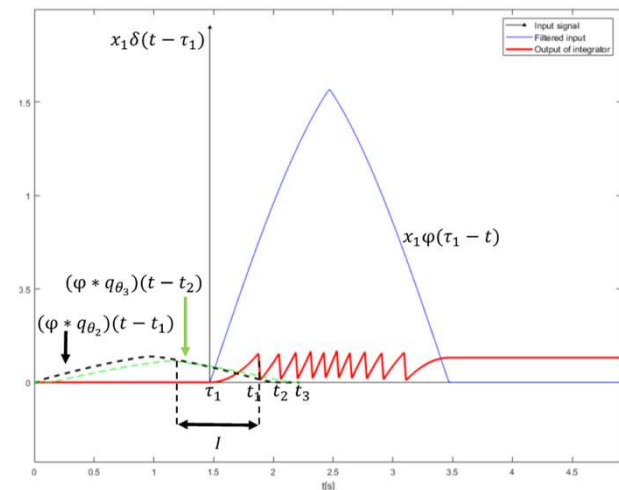
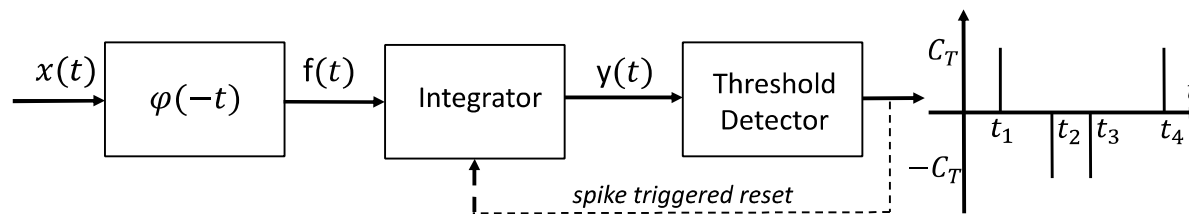
(d)

# Summary on Sparse Sampling with Comparator

- We can sample and perfectly reconstruct non-bandlimited signals 👍
- Number of time samples still large (time information provided also when signal is zero) 🗨️
- Use the new framework but with the Integrate-and-Fire TEM 👍 👍 👍

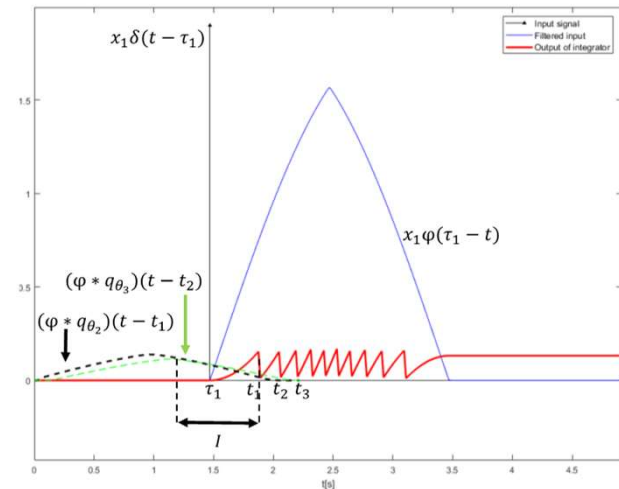
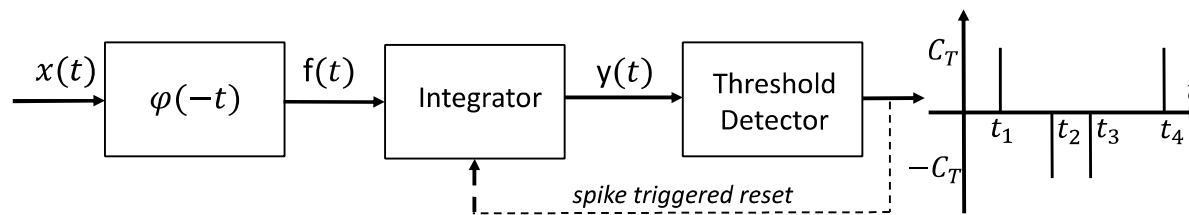


# Integrate and Fire TEM



- The sampling kernel  $\varphi(t)$  and its non-uniform shifts reproduce  $e^{j\omega_0 t}$  and  $e^{-j\omega_0 t}$  and  $0 < \omega_0 \leq \frac{\pi}{L}$  where  $L$  is the support of  $\varphi(t)$ .
- What is the minimum value of the trigger mark  $C_T$  that would allow the perfect reconstruction of stream of pulses or piecewise constant signals?

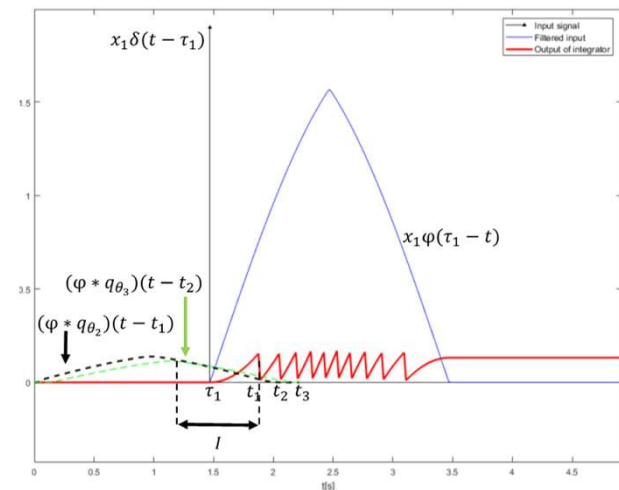
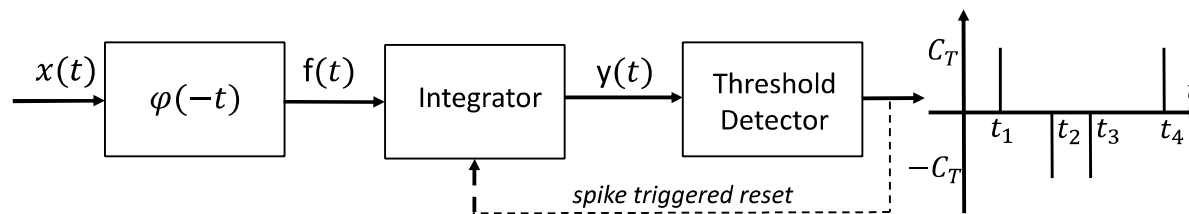
# Integrate and Fire TEM



- Given the times  $t_1, t_2, \dots, t_n$ , the amplitude values are

$$y_n = y(t_n) = \pm C_T = \int_{t_{n-1}}^{t_n} f(\tau) d\tau = \int_{t_{n-1}}^{t_n} \int x(\alpha) \phi(\alpha - t) d\alpha d\tau.$$

# Integrate and Fire TEM



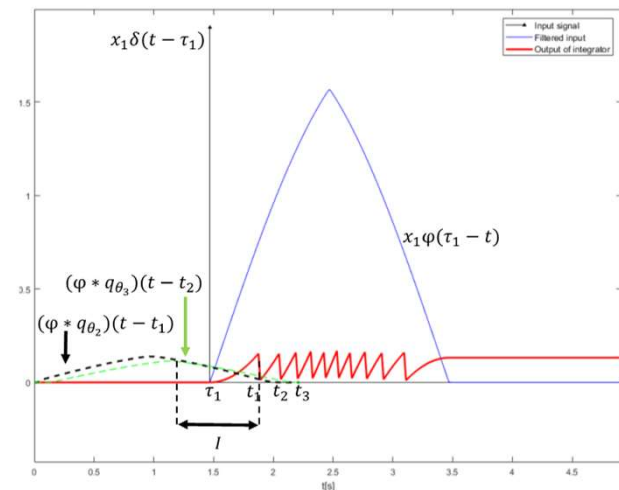
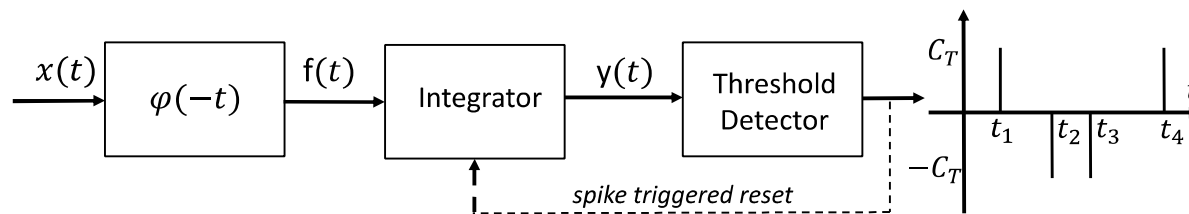
- Equivalently the output samples can be expressed as:

$$y(t_n) = \langle x(t), (\phi * q_{\theta_n})(t - t_{n-1}) \rangle,$$

where  $\theta_n = t_n - t_{n-1}$  and  $q_{\theta_n}(t)$  is defined as:

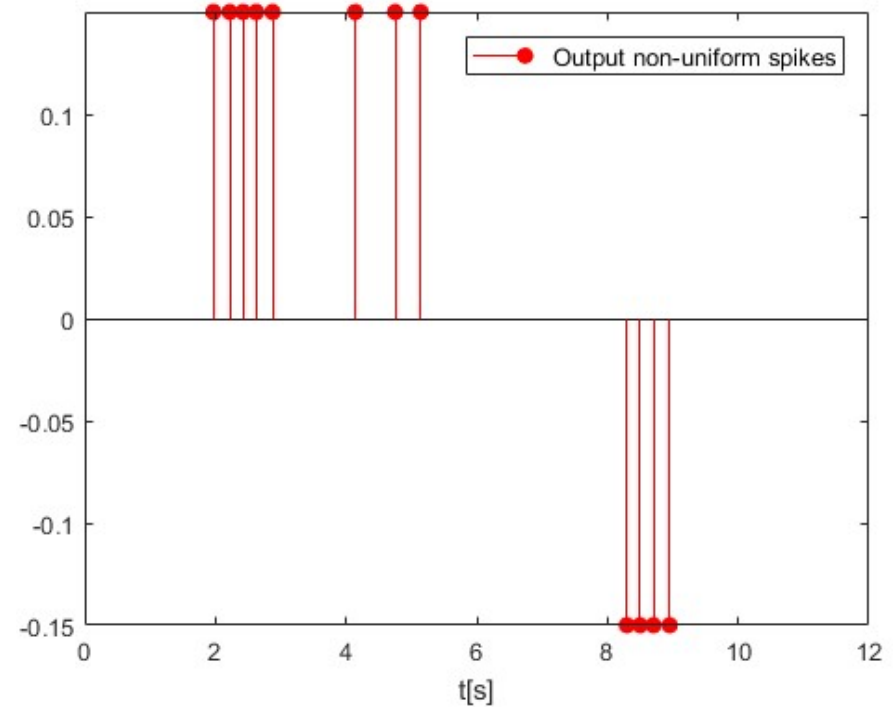
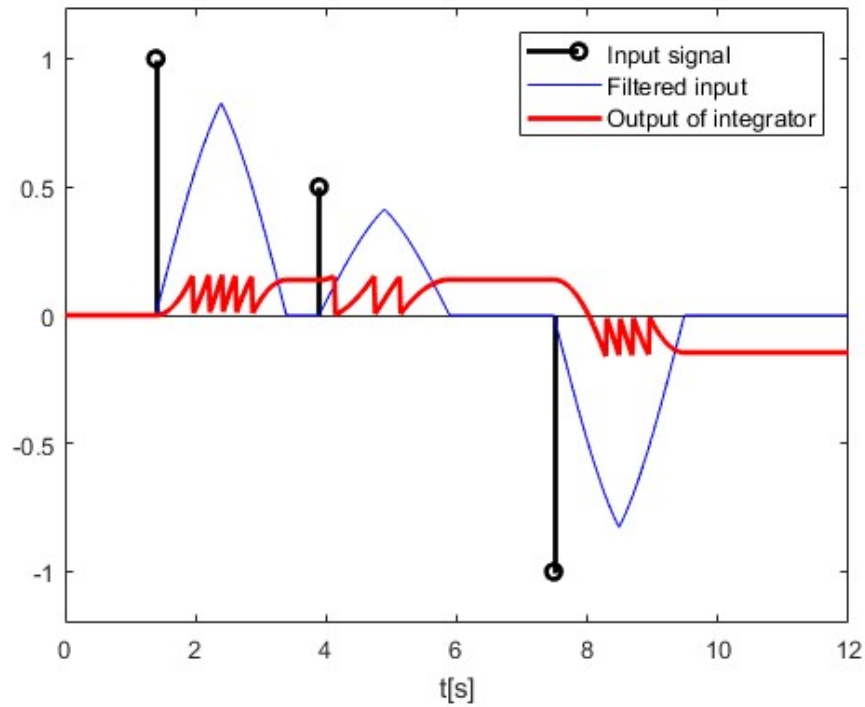
$$q_{\theta_n}(t) = \begin{cases} 1, & 0 \leq t \leq \theta_n, \\ 0, & \text{otherwise.} \end{cases}$$

# Integrate and Fire TEM

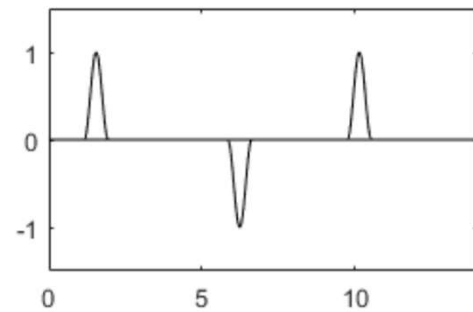


- The equivalent kernel  $(\varphi * q_{\theta_n})(t - t_{n-1})$  is still able to reproduce exponentials
- So trigger mark must guarantee enough samples in a short interval
- **Proposition:** when  $C_T < \frac{A_{min}}{4\omega_0^2} \left(1 - \cos\left(\frac{\omega_0 L}{2}\right)\right)$  then  $t_1, t_2, t_3 \in \left[\tau_1, \tau_1 + \frac{L}{2}\right]$  and perfect reconstruction is possible

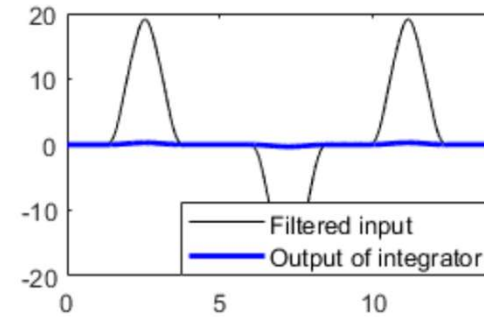
# Integrate and Fire – Reconstruction of Pulses



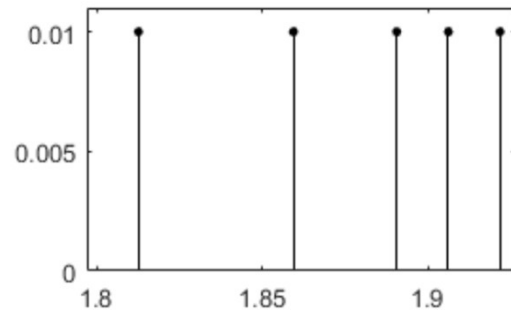
# Integrate and Fire – Reconstruction of Pulses



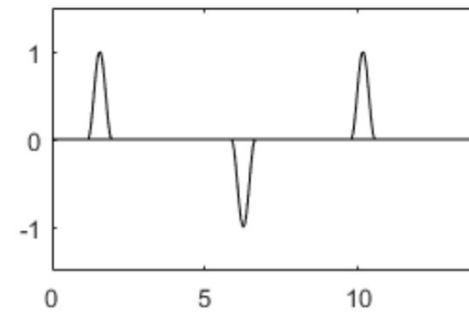
(a)



(b)



(c)



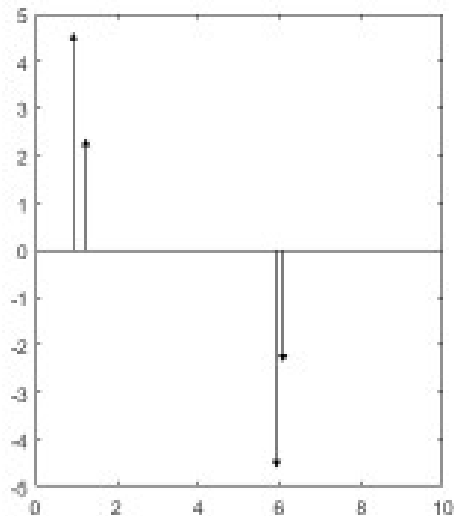
(d)



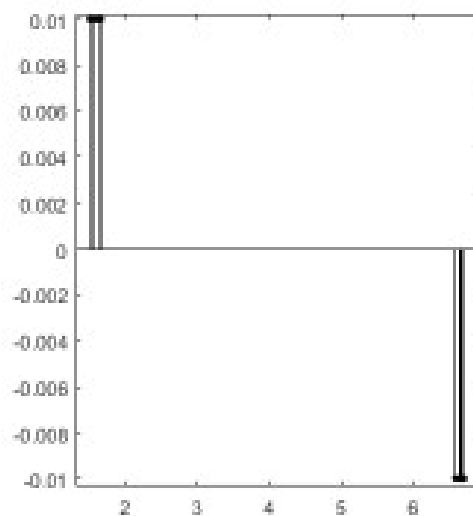
## Reconstruction with Arbitrary Kernels

- Sufficient conditions for perfect reconstruction may appear restrictive, but they can be relaxed with minimum loss in reconstruction quality
- The proposed reconstruction framework can be used with *any* acquisition device
- If reproduction of exponentials is not satisfied use LS methods to find the coefficients  $c_{m,n}^l$  that achieve best fit:

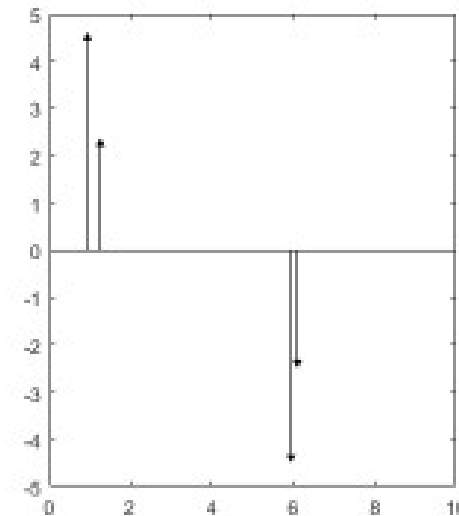
$$\langle f(t), \phi(t - t_n) \rangle = \sum_{k=1}^N c_{m,k}^l \langle \phi(t - t_k), \phi(t - t_n) \rangle, \text{ so that } \sum_{n=1}^N c_{m,n}^l \phi(t - t_n) \approx e^{j\omega_m t},$$



(a)

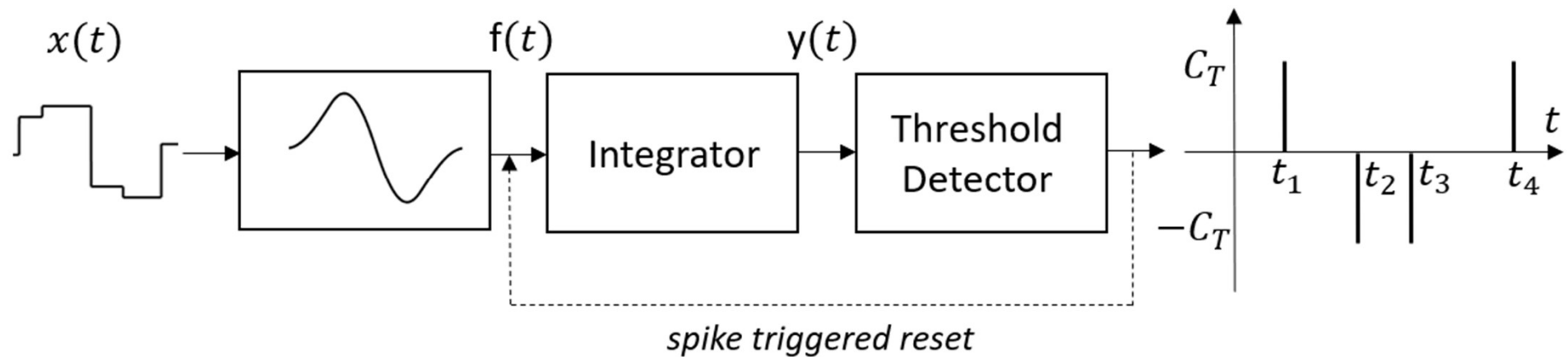


(b)



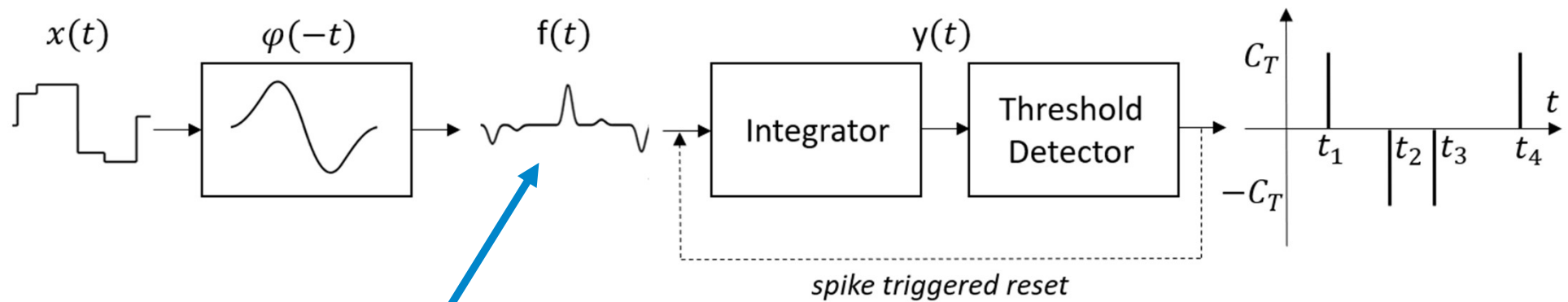
(c)

# Integrate and Fire – Piecewise Constant Signals



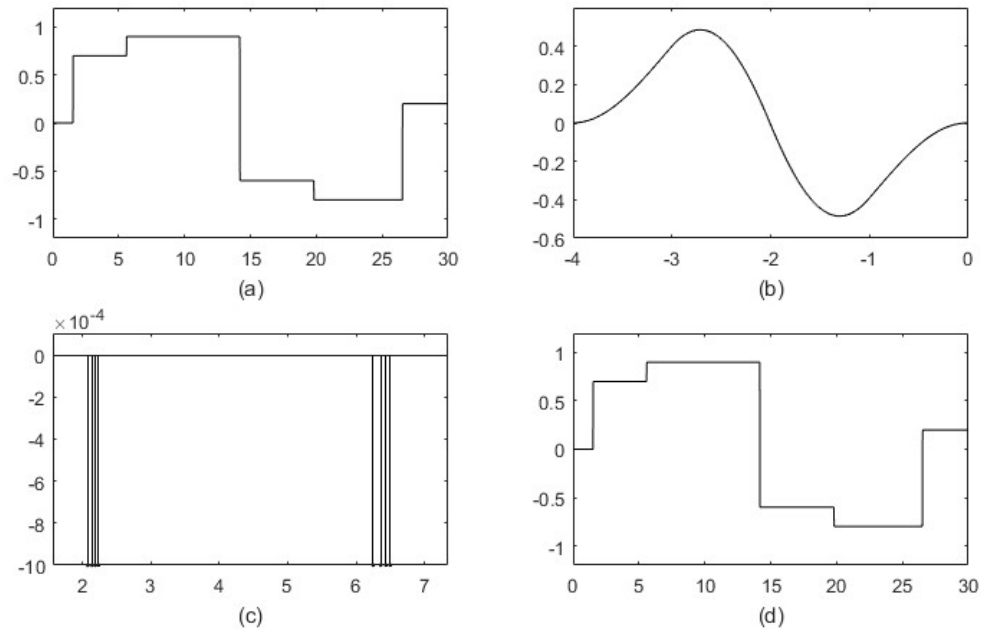
This is equivalent to the way a pixel operates  
in neuromorphic video cameras

# Integrate and Fire – Piecewise Constant Signals



Filtered Stream of Diracs

## Integrate and Fire – Piecewise Constant Signals



If the distance  $S$  between discontinuities is on average  $S > (L - 1)T$  with  $T$  being the sampling period in uniform sparse sampling<sup>4</sup> then our time encoding framework is **more efficient** than uniform sampling (lower sampling density) 👍👍👍

<sup>4</sup>P.L. Dragotti, M. Vetterli and T. Blu, Sampling Moments and Reconstructing Signals of Finite Rate of Innovation: Shannon meets Strang-Fix, IEEE Trans. on Signal Processing, vol.55 (5), pp. 1741-1757, May 2007.

## Spike-Based Processing

- Sensing efficiently is only half of the story
- Once a signal has been converted into spikes, how do we process it efficiently?
  - **Creating an AI can be five times worse for the planet than a car** (resource NewScientist)
  - How do we compute fundamental transforms (e.g., Fourier or Wavelet Transforms)
  - Can we find the sparse representation of a signal using spiking neuron models? (Some results based on spike rates<sup>4,5</sup>)
  - Deep learning with spiking signals?<sup>6</sup>

<sup>4</sup>P.T.P. Tang, T.-H. Lin, and M. Davies, “Sparse coding by spiking neural networks: Convergence theory and computational results” arXiv:1705.05475 , 2017.

<sup>5</sup>C. Pehlevan, “A Spiking Neural Network with Local Learning Rules Derived From Nonnegative Similarity Matching”, ICASSP 2019.

<sup>6</sup>E. Neftci, “Surrogate Gradient Learning in Spiking Neural Networks,”, arXiv:1901.09948, 2019.

## Conclusions

- **Event-based sensing and processing** is an emerging and exciting research area!
- Topic at the **intersection** of signal processing, computational neuroscience and machine learning
- Proved sufficient conditions for the exact reconstruction of classes of sparse signals from time-based information
- Many open questions on both the sensing and the processing front
  - Multi-dimensional case
  - Adaptive acquisition
  - $L_1$  optimization strategies
  - Learning sparsifying representations for spiking signals

## References

[1] Roxana Alexandru and Pier Luigi Dragotti, "Reconstructing classes of non-bandlimited signals from time encoded information", Available online at arXiv:1905.03183.

[2] Roxana Alexandru and Pier Luigi Dragotti, "Time encoding and perfect recovery of non-bandlimited signals with an integrate-and-fire system.",, International Conference on Sampling Theory and Applications, Bordeaux, France.

[3] Roxana Alexandru and Pier Luigi Dragotti "Time-based sampling and reconstruction of non-bandlimited signals", IEEE International Conference on Acoustics, Speech and Signal Processing, Brighton, United Kingdom.

**Thank you.**