# **Signal Processing Project - Team 17**

# Sampling Signals with Finite Rate of Innovation

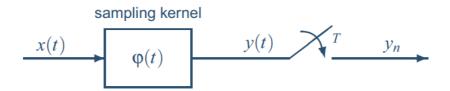
Agrim Rawat - 2020102037 Aditya Nair - 2020102022

### 1. Abstract

We define rate of innovation as the number of degrees of freedom that a class of signal has, such as a stream of Diracs, nonuniform splines, and piecewise polynomials. In this paper, we show that we can uniformly sample these signals at the rate of innovation, or even higher, by using an appropriate kernel, allowing us to perfectly reconstruct these signals even though they are not bandlimited. We basically sample and reconstruct periodic and finite–length stream of Diracs, nonuniform splines, and piecewise polynomials, using sinc and Gaussian kernels. For infinite–length signals, we define a *local* finite rate of innovation, which we sample and reconstruct based on spline kernels. We identify the innovative part of a signal, such as the time instants and the weight of Diracs, using an annihilating filter or a locator filter.

# 2. General Sampling Setup

The general sampling setup involves convolving the original signal x(t) with a sampling kernel  $\varphi(t)$  to get a filtered signal y(t) which is further sampled at intervals T to get the samples  $y_n = y(nT)$ .



If there is no sampling kernel, we simply get y(nT) = x(nT).

### 2.1 Shannon Reconstruction for Bandlimited Signals

# Time domain: $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $\sum_{n \in \mathbb{Z}} \delta(t-nT)$ Frequency domain: $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$ $x(t) \longrightarrow x(t) = \sum_{n \in \mathbb{Z}} y_n sinc(t-nT)$

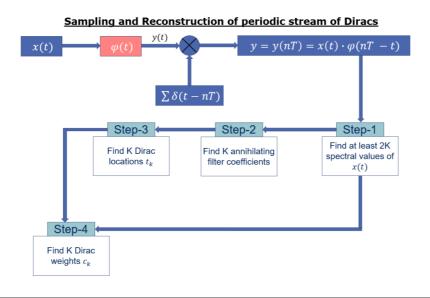
If x(t) is bandlimited, or  $X(\omega)=0 \ \forall \ \omega>\omega_m$ , then samples x(nT) with  $\omega_s\geq 2\omega_m \implies T\leq \pi/\omega_m$  (Nyquist Condition) are sufficient to reconstruct x(t).

Calling  $B=2\omega_m/2\pi$ , which is the bandwidth of x(t) in cycles per second, we see that B=1/T sample are a sufficient representation of x(t). The reconstruction formula is given by

$$x(t) = \sum_{n=-\infty}^{\infty} x(nT) \mathrm{sinc}\left(rac{t}{T} - n
ight)$$

### 2.2 Non-Bandlimited Case

To generalize the Shannon Reconstruction to non-bandlimited signals we consider signals with finite rate of innovation.



# 3. Signals with Finite Rate of Innovation

Definition: A signal with a finite rate of innovation is a signal with parametric representation as

$$x(t) = \sum_{n \in \mathbb{Z}} \sum_{r=0}^{R} c_{nr} arphi_r \left(rac{t-t_n}{T}
ight)$$

for a set of functions  $\{ arphi_r(t) \}_{r=0,...R}$  with a finite rate of innovation ho defined as

$$ho = \lim_{ au o \inf} rac{1}{ au} C_x(rac{- au}{2},rac{ au}{2})$$

- If we consider finite length signals, or periodic signals with length  $\tau$  then degrees of freedom is finite, and then rate of innovation will be  $\frac{1}{\tau}C_x(0,\tau)$ .
- ullet For infinite length signals, we define *local* rate of innovation with a moving window of size au as

$$ho_T(t) = rac{1}{ au} C_x (t - rac{ au}{2}, t + rac{ au}{2})$$

# 4. Periodic Continuous-Time case

We represent such signals as

$$x(t) = \sum_{m \in \mathbb{Z}} X[m] e^{i(2\pi mt/ au)}$$

### 4.1 Stream of Diracs

We take a stream of Diracs with a period of au, and a rate of innovation as

$$\rho = \frac{2K}{\tau}$$

The stream of Diracs can be written as

$$egin{align} x(t) &= \sum_{m \in \mathbb{Z}} c_m \delta(t-t_n) \ &= \sum_{m \in \mathbb{Z}} \underbrace{rac{1}{ au} igg( \sum_{k=0}^{K-1} c_k e^{-i(2\pi m t_k/ au)} igg)}_{X[m]} e^{i(2\pi m t/ au)} \end{aligned}$$

where X[m] is the Fourier series coefficients. Taking another Fourier series  $A[m], m \in \{0, 1, \ldots, K\}$  with z-transform,

$$A(z)=\sum_{m=0}^K A[m]z^{-m}$$

and taking the zeros of the Fourier transform at  $u_k = e^{(-i(2\pi t_k/ au)}$  gives us

$$A(z) = \prod_{k=0}^{K-1} \left(1 - e^{(-i(2\pi t_k/ au)} z^{-1}
ight)$$

Since X[m] is the sum of K exponentials, and each of them become zero by one of the roots of A[m], we can deduce that

$$\boxed{A[m]*X[m]=0}$$

Here, A[m] is an <u>annihilating filter</u>. We obtain the time-domain function x(t) by performing inverse Fourier-Transform.

Let us define a sinc function of bandwidth  $[-B\pi, B\pi], B \in \mathbb{R}_+$  and name it as

$$h_B(t) = B \cdot sinc(Bt)$$

**Theorem 1:** Consider x(t) as a periodic stream of Diracs, with a period of  $\tau$ , with K Diracs of weight  $\{c_k\}_{k=0}^{K-1}$  and at location  $\{t_k\}_{k=0}^{K-1}$ . Taking sampling kernel as the sinc function stated above, where  $B \geq \rho$ , and sampling  $(h_B * x)(t)$  at N uniformly-distributed locations  $t = nT, n \in \{0, 1, \ldots N-1\}, N \geq 2M+1, M = \lfloor \frac{B_\tau}{2} \rfloor$ , then the samples that we collect, i.e.

$$y_n = \langle h_B(t-nT), x(t) \rangle, \ n = 0, \dots, N-1$$

are sufficient to reconstruct the original signal x(t)

*Proof:* We follow the steps as follows:

1. We first find X[m],  $|m| \leq M$  from our samples  $y_n$ . We plug in the general form of x(t) in the above equation, giving us

$$y_n = \sum_{m=-M}^M X[m] \cdot e^{i(2\pi m n T/ au)}$$

- When  $\tau = \lambda T$ , i.e. when  $\tau$  is divisible by T, we observe that this simply is the IDTFT of X[m].
- 2. We then find the annihilating coefficients for X[m] by solving the equation  $A[m]\ast X[m]=0$ . We write an equivalent form of this as

$$\underbrace{ \begin{bmatrix} X[0] & X[-1] & \cdots & X[-K+1] \\ X[1] & X[0] & \cdots & X[-K+2] \\ \vdots & \vdots & \ddots & \vdots \\ X[K-1] & X[K-2] & \cdots & X[0] \end{bmatrix} \cdot \begin{bmatrix} A[1] \\ A[2] \\ \vdots \\ A[K] \end{bmatrix} = - \begin{bmatrix} X[1] \\ X[2] \\ \vdots \\ X[K] \end{bmatrix} }_{X[K]}$$

Yule-Walker system

- Here, we obtain a unique solution for the case when there are K distinct Diracs in x(t).
- 3. We factorize A(z) into its roots as

$$A(z) = \prod_{k=0}^{K-1} \left(1 - u_k z^{-1}
ight)$$

where  $\,u_k=e^{(-i(2\pi t_k/ au)}$  , which gives us the locations of  $\{t_k\}_{k=0}^{K-1}$ 

4. We then find the weights of the Diracs  $c_k$  by using the value of  $u_k$  as follows, by using  $X[m]=rac{1}{ au}\sum_{k=0}^{K-1}c_k\cdot e^{-i(2\pi mt_k/ au)}$ :

$$\begin{bmatrix} X[0] \\ X[1] \\ \vdots \\ X[K-1] \end{bmatrix} = \frac{1}{\tau} \begin{bmatrix} 1 & 1 & \cdots & 1 \\ u_0 & u_1 & \cdots & u_{K-1} \\ \vdots & \vdots & \ddots & \vdots \\ u_0^{K-1} & u_1^{K-1} & \cdots & u_{K-1}^{K-1} \end{bmatrix} \cdot \begin{bmatrix} c_0 \\ c_1 \\ \vdots \\ c_{K-1} \end{bmatrix}$$

Vandermonde System

• Here, we obtain a unique solution for the case when all the  $t_k$ 's are different.

## 4.2 Nonuniform Splines

A signal x(t) is a periodic nonuniform spline of degree R with discontinuities at  $\{t_k\}_{k=0}^{K-1} \in [0,\tau]$  iff its  $(R+1)^{th}$  derivative is a periodic stream of K weighted Diracs, i.e.

$$x^{(R+1)}(t) = \sum_{n \in \mathbb{Z}} c_n \delta(t-t_n), \; t_{n+K} = t_n + au, \; c_{n+K} = c_n orall \; n \in \mathbb{Z}$$

If we differentiate the standard form of the signal x(t) R + 1 times, we see that the Fourier series coefficients are

$$X^{R+1}[m] = \left(rac{i2\pi m}{ au}
ight)^{R+1} \cdot X[m], \ m \in \mathbb{Z}$$

From the previously stated theorem, we can obtain the stream of  ${\it K}$  Diracs from the above coefficients, and thus reconstruct the periodic nonuniform spline.

**Theorem 2:** If we take a periodic nonuniform spline x(t) periodic with period  $\tau$  and containing K pieces of maximum degree R, and take samples of  $(h_B*x)(t)$  at N uniform locations  $t=nT,\ n=0,\ldots,N-1,\ N\geq 2M+1,\ M=\lfloor B_\tau/2\rfloor$ . Then we can say that the samples

$$y_n = \langle h_B(t-nT), x(t) 
angle$$

are sufficient to reconstruct the original signal x(t).

### 4.3 Derivative of Diracs

We take the  $r^{th}$  derivative of the Dirac function, which has the property as follows:

$$\int f(t)\cdot\delta^{(r)}(t-t_0)dt=(-1)^r\cdot f^{(r)}(t_0)$$

here, f(t) is r times continuously differentiable. We take a periodic stream of differentiated Diracs

$$x(t) = \sum_{n \in \mathbb{Z}} \sum_{r=0}^{R_n-1} c_{nr} \delta^{(r)}(t-t_n)$$

here the degrees of freedom is different than usual. Here we have K locations, but our weights are different this time, i.e. our weights are  $\tilde{K} = \sum_{k=0}^{K-1} R_k$ , which gives us degrees of freedom as  $K + \tilde{K}$ .

We find the corresponding Fourier series coefficients as

$$X[m] = \underbrace{rac{1}{ au}\sum_{k=0}^{K-1}\sum_{r=0}^{R_k-1}c_{kr}\left(rac{i2\pi m}{ au}
ight)}_{ ilde{c}_{kr}}e^{-i(2\pi mt_k/ au)}$$

We see that the filter

$$A(z) = \prod_{k=0}^{K-1} (1-u_k z^{-1})^{R_k}$$

has  $R_k$  poles, at  $z=u_k$  and annihilates X[m]. We find the locations of  $t_k$  in the same method as previously, stated, but here the weights  $\tilde{c}_{kr}$  are found by solving the above Fourier series coefficients for  $m=0,\ldots,\tilde{K}-1$ .

**Theorem 3:** If we take a stream of differentiated Diracs x(t) periodic with period  $\tau$ , and take samples of  $(h_B * x)(t)$  at N uniform locations, with the same parameters as Theorems 1 and 2, then the samples

$$y_n = \langle h_B(t-nT), x(t) \rangle, \ n = 0, \dots, N-1$$

### 4.4 Piecewise Polynomials

We define piecewise polynomials in a fashion similar to the periodic nonuniform splines, where its  $(R+1)^{th}$  derivative must be a stream of differentiated Diracs. Here, d.o.f. will be K from the locations, and  $\tilde{K}=(R+1)K$ , thus we get

$$\rho = \frac{(R+2)K}{\tau}$$

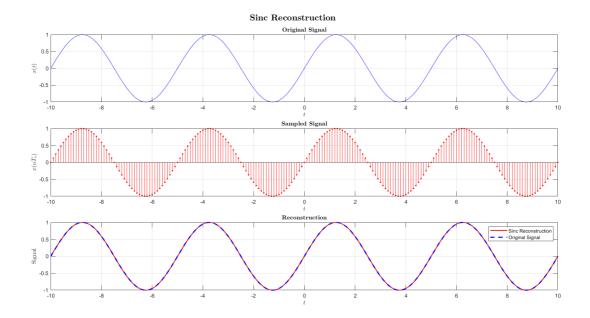
**Theorem 4:** If we take a piecewise polynomial x(t) periodic with period  $\tau$ , having K pieces of maximum degree R and take samples of  $(h_B*x)(t)$  at N uniform locations, with the same parameters as Theorems 1 and 2, then the samples

$$y_n = \langle h_B(t-nT), x(t) \rangle, \; n = 0, \ldots, N-1$$

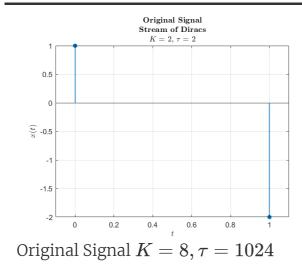
is sufficient to reconstruct x(t).

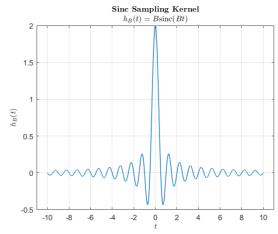
# 5. Results and Inferences

### **Bandlimited case**

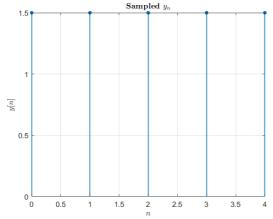


### Non-bandlimited case

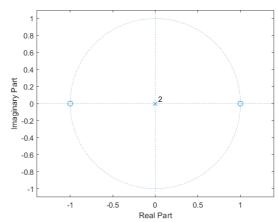




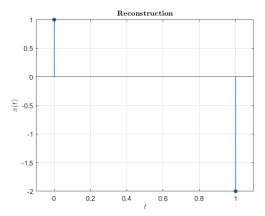
The sampling kernel



The sampled signal y(nT)



The zeros of the annihilating filter



Reconstructed Signal

We see that we get a perfect reconstruction of the original signal.  $\label{eq:construction}$ 

# 6. References

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- 4. 'Sampling curves with finite rate of innovation', Hanjie Pan, Thierry Blu, Pier-Luigi Dragotti.