

Maximal Gap of a Sampling Set for the Exact Iterative Reconstruction Algorithm in Shift Invariant Spaces

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Abstract—A conventional A/D converter prefilters a signal by an ideal lowpass filter and performs sampling for bandlimited signals by the Nyquist sampling rate. Recent research reveals that A/D conversion in a shift invariant space provides more flexible choices for designing a practical A/D conversion system of high accuracy. This paper focuses on the maximal gap of a sampling set for the iterative algorithm in shift invariant spaces, which provides an explicit formula to calculate the maximal gap of a sampling set in terms of a generator of the undertaken shift invariant spaces.

Index Terms—A/D conversion, maximal gap, nonuniform sampling, prefilter, shift invariant space, sampling set.

I. INTRODUCTION

IN digital signal processing and digital communications, an analog signal is converted into a digital signal by an A/D (analog-to-digital) conversion device.

A signal f is said to be of *finite energy* if $\|f\| < \infty$, where $\|\cdot\|$ is the square norm defined by $\|f\| = (\int_{\mathbb{R}} |f(t)|^2 dt)^{1/2}$. We also denote by $L^2(\mathbb{R})$ the collection of all signals of finite energy, that is, $\{f: \|f\| < \infty\}$. f is said to be *bandlimited* if $\hat{f}(\omega) = 0$ whenever $|\omega| > \sigma$ for some $\sigma > 0$, where \hat{f} is the *Fourier transform* of f defined by $\hat{f}(\omega) = \int_{\mathbb{R}} f(t)e^{-it\omega} dt$. In this case, f is also called a σ -band signal.

A conventional A/D converter prefilters a signal of finite energy by an ideal lowpass filter and performs uniform sampling by the Nyquist sampling rate [33]. Since in some practical systems sampling cannot be always made uniformly, one has to consider the nonuniformly sampled signal [6], [20].

A discrete set $X = \{t_k\}_k \subset \mathbb{R}$ is called a *sampling set* for a signal f if f is completely determined by the sample set $f(X) = \{f(t_k)\}_k$; X is said to be *separated* if $\inf_{k \neq j} |t_k - t_j| = d > 0$, where d is called the *separation* of X . A sampling set $X = \{t_k\}_k$ is said to be *stable* for a signal space S if there is a constant $C \geq 1$ such that

$$C^{-1}\|f\| \leq \left(\sum_k |f(t_k)|^2 \right)^{1/2} \leq C\|f\| \quad (1)$$

holds for any signal $f \in S$.

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The nonuniform sampling for the bandlimited signals is studied by Beurling, Landau and others [6], [23], [29]. It is understood that a separated X is a stable sampling set for the σ -band signal space if the *Beurling lower density*¹ $D^-(X) > \pi/\sigma$. Conversely, $D^-(X) \geq \pi/\sigma$ if X is a stable sampling set for the σ -band signal space. Since Beurling-Landau's theorem does not provide a reconstruction formula, Feichtinger and Gröchenig established an iterative reconstruction algorithm to handle the nonuniformly sampled bandlimited signals [20].

For a $\lambda \geq 1$, the *shift invariant space* $V_\lambda(\varphi)$ generated by a function φ is defined by [8], [26]

$$V_\lambda(\varphi) = \left\{ \sum_k c_k \varphi(\lambda \cdot -k): \sum_k |c_k|^2 < \infty \right\}. \quad (3)$$

The φ and λ are, respectively, called a *generator* and the *dilation* of the shift invariant space $V_\lambda(\varphi)$. Let $\text{sinc} = (\sin \pi t / \pi t)$. Then $V_\lambda(\text{sinc})$ contains exactly all $\pi\lambda$ -band signals of finite energy. Then the conventional A/D conversion can be formulated as A/D conversion in $V_\lambda(\text{sinc})$. One can therefore consider A/D conversion in a general shift invariant space $V_\lambda(\varphi)$ for a general generator φ .

In real world application, such an extension of A/D conversion is useful and necessary, e.g., for performing nonideal prefiltering [34], for avoiding Gibbs phenomenon in FFT [17], for using the impulse response of fast decay [32], for taking into account real acquisition and reconstruction devices [38], for considering arbitrary band signals [24], for obtaining smoother frequency cutoff or for numerical implementation [4], [5], [37], [38], [40]. This is formulated by choosing an appropriate function φ with some desirable shape corresponding to a particular "impulse response" of a device, such as a compactly supported function, a function with polynomial or exponential decay, or a function φ with smooth cutoff frequency $\hat{\varphi}$. Then one prefilters a signal by the *shift invariant space* $V_\lambda(\varphi)$ [7], [11], [37], and performs sampling in $V_\lambda(\varphi)$ [5], [10], [12], [14], [15], [37], [39]. Furthermore, A/D conversion in shift invariant spaces does provide more flexible choices to design various A/D conversion systems of high accuracy [5], [10]–[15], [36], [37], [39], [40].

The prefiltering, uniform sampling and oversampling has been well handled in the shift invariant spaces [7], [11]–[13], [39], [27], [37], [40]. However, the nonuniform sampling in shift invariant spaces is relatively tough. For instance, the

¹Let $v^-(r)$ denote the minimum number of points of X to be found in the interval $(-r, r)$, formally, $v^-(r) = \min_{x \in \mathbb{R}} \#(X \cap (x - r, x + r))$. The Beurling lower density is defined by

$$D^-(X) = \lim_{r \rightarrow \infty} \frac{v^-(r)}{2r}. \quad (2)$$

Beurling-Landau type sampling theorem in shift invariant spaces is not yet established so far [3]. In 1996, Liu [31] established the iterative reconstruction algorithm [20], [22] in the spline shift invariant spaces, and then, in 1998, Aldroubi and Feichtinger [2] established the iterative reconstruction algorithm in a general shift invariant space. But the maximal gap of the a sampling set for the iterative algorithm in a general shift invariant space is unknown so far. Our objective in this paper is to find the maximal gap of sampling set for the iterative reconstruction algorithm in a general shift invariant space. We shall also calculate some typical examples to illustrate our result.

II. EXACT ITERATIVE RECONSTRUCTION ALGORITHM IN SHIFT INVARIANT SPACES AND THE MAXIMAL GAP OF A SAMPLING SET

To prefilter a signal f by a shift invariant space $V_\lambda(\varphi)$ is equivalent to making a *quasiprojection* $P: L^2(\mathbb{R}) \rightarrow V_\lambda(\varphi)$ defined by

$$P(f) = \sum_k \langle f, \varphi(\cdot - k) \rangle \varphi(\cdot - k)$$

where $\langle \cdot, \cdot \rangle$ is the *inner product* in $L^2(\mathbb{R})$ defined by

$$\langle f, g \rangle = \int_{\mathbb{R}} f(t)g(t) dt, \quad f, g \in L^2(\mathbb{R}).$$

The aliasing error $e = \|f - P(f)\|$ can be made arbitrarily small as long as the dilation λ is sufficiently large [7], [11], [27], [28]. In this paper, we focus on the nonuniform sampling in shift invariant spaces. Hence we shall ignore the prefiltering and assume that the signal f is taken from a shift invariant space $V_\lambda(\varphi)$. In this section, we are going to find the maximal gap of sampling set for the iterative algorithm in shift invariant spaces. We shall work in the framework of *Wiener amalgam spaces* [19], [21], which is commonly used in sampling theorem for shift invariant spaces. A Wiener amalgam space W^p for some $p \geq 1$ consists of all measurable functions φ for which the norm

$$\|\varphi\|_{W^p} = \left(\sum_k \sup_{t \in [0,1]} |\varphi(t - k)|^p \right)^{1/p} < \infty.$$

We now introduce the exact iterative algorithm [20] in shift invariant spaces in a different sense to [1], [2].

For a discrete set $X = \{t_k\}_k$, the δ -ball $B_\delta(t_k)$ for a $\delta > 0$ is defined by

$$B_\delta(t_k) = \{t: |t_k - t| \leq \delta\}. \quad (4)$$

X is called δ -dense if $\cup_k B_\delta(t_k) = \mathbb{R}$, where δ is called the *maximal gap*² of X if δ is the smallest one such that X is δ -dense. Take a sequence $\{U_k\}_k$ of nonnegative functions such that U_k is supported in $B_\delta(t_k)$ and

$$\sum_k U_k = 1. \quad (5)$$

²Maximal gap in this paper does not mean the ‘‘optimal maximal gap,’’ which is the maximum one among the gaps between two consecutive samples of X , i.e., $\delta = \sup_k |t_k - t_{k-1}|$ if $X = \{t_k\}_k$.

Then the $\{U_k\}_k$ is called a *bounded uniform partition of unity (BUPU)* with respect to X . For a signal $f \in W^2$, we define the interpolation operator S_X by

$$S_X(f) = \sum_k f(t_k)U_k. \quad (6)$$

It is well-known that the operator S_X maps the space W^2 into the space $W^2 \subset L^2(\mathbb{R})$ if X is separated [18]. We also use the symbol G_φ with the norms $\|\cdot\|_0$ and $\|\cdot\|_\infty$ being defined by

$$G_\varphi = \left(\sum_k |\hat{\varphi}(\cdot + 2k\pi)|^2 \right)^{1/2} \quad (7)$$

$$\|G_\varphi\|_0 = \inf_{\omega \in [0, 2\pi]} G_\varphi(\omega) \quad (8)$$

$$\|G_\varphi\|_\infty = \sup_{\omega \in [0, 2\pi]} G_\varphi(\omega). \quad (9)$$

For a signal $f \in V_\lambda(\varphi)$, it is easy to show $f \in W^2$ if $\varphi \in W^1$ [2], [10], [18]. Assume $\varphi \in W^1$. Then we can interpolate the samples $\{f(t_k)\}_k$ of a signal $f \in V_\lambda(\varphi)$ with $S_X(f)$. By projecting the interpolation function $S_X(f)$ into $V_\lambda(\varphi)$, we get an approximation $f_1 = PS_X(f) \in V_\lambda(\varphi)$ of the original signal f . Then we interpolate the samples $\{E_1(t_k) = f(t_k) - f_1(t_k)\}_k$ of the error $E_1 = f - f_1 \in V_\lambda(\varphi)$ with $S_X(E_1)$. By projecting the interpolation function $S_X(E_1)$ of the error E_1 , we get an approximation $e_1 = PS_X(E_1) \in V_\lambda(\varphi)$ of the error E_1 . Adding e_1 to f_1 , we get a new approximation $f_2 = f_1 + e_1$ to the original signal $f \in V_\lambda(\varphi)$. Let $E_2 = f - f_2$, and repeat the interpolating and projecting procedures. We obtain an approximation sequence $f_{n+1} = f_1 + \sum_{k=1}^n e_k$ of the original signal f . In the following we will show that f_n converges to f in $L^2(\mathbb{R})$.

By the algorithm defined above, we know $E_n = f - f_n$. Then $E_{n+1} - E_n = f_n - f_{n+1} = -e_n = -PS_X(f - f_n) = -PS_X(E_n)$, that is $E_{n+1} = E_n - PS_X(E_n)$. Let $T = I - PS_X$, where I is the identity operator. Then $E_{n+1} = T(E_n)$, and consequently $\|E_{n+1}\| \leq \|T\| \|E_n\|$ where $\|T\|$ is the norm of the operator T defined by

$$\|T\| = \sup_{f \in L^2(\mathbb{R})} \frac{\|T(f)\|}{\|f\|}.$$

If we can show that T is a contraction, that is, $\|T\| < 1$. Then $\|E_n\| \rightarrow 0$. Consequently $f_n \rightarrow f$ in $L^2(\mathbb{R})$.

In order to show that T is a contraction, we define the oscillation function $\text{osc}_\delta(f)$ of a signal f as

$$\text{osc}_\delta(f) = \sup_{|s| \leq \delta} |f - f(\cdot + s)|. \quad (10)$$

For a signal $f \in V_\lambda(\varphi)$, there is a $\{c_k\}_k \in l^2$ such that $f = \sum_k c_k \varphi(\lambda \cdot - k)$. Therefore,

$$\text{osc}_\delta(f)(t) \leq \int_t^{t+\delta} |f'(s)| ds = \int_{\mathbb{R}} |f'(s+t)| \chi_{[0,\delta]}(s) ds.$$

Then

$$\|\text{osc}_\delta(f)\| \leq \int_{\mathbb{R}} \|f'(s + \cdot)\| \chi_{[0,\delta]}(s) ds = \delta \|f'\|.$$

By Parseval identity, we have

$$\begin{aligned}
\|f'\| &= \lambda \left\| \sum_k c_k \varphi'(\lambda \cdot - k) \right\| \\
&= \frac{\lambda}{\sqrt{2\pi}} \left\| \frac{1}{\lambda} \hat{\varphi}'(\cdot/\lambda) \sum_k c_k e^{-ik\cdot/\lambda} \right\| \\
&= \frac{\lambda}{\sqrt{2\pi}} \left\| \sqrt{\frac{1}{\lambda}} \hat{\varphi}' \sum_k c_k e^{-ik\cdot} \right\| \\
&= \frac{\lambda}{\sqrt{2\pi}} \sqrt{\int_0^{2\pi} \left| \frac{G_{\varphi'}(\omega)}{\sqrt{\lambda} G_{\varphi}(\omega)} G_{\varphi}(\omega) \sum_k c_k e^{-ik\cdot} \right|^2 d\omega} \\
&\leq \lambda \left\| \frac{G_{\varphi'}}{G_{\varphi}} \right\|_{\infty} \|f\|.
\end{aligned}$$

Therefore,

$$\|\text{osc}_{\delta}(f)\| \leq \delta \lambda \left\| \frac{G_{\varphi'}}{G_{\varphi}} \right\|_{\infty} \|f\|. \quad (11)$$

Now we have to find the condition on δ such that T is a contraction

$$\begin{aligned}
\|T(f)\| &= \|f - PS_X(f)\| \\
&= \|P(f) - PS_X(f)\| \\
&\leq \|P\| \|f - S_X(f)\| \\
&\leq \|P\| \|\text{osc}_{\delta}(f)\| \\
&\leq \lambda \delta \|P\| \left\| \frac{G_{\varphi'}}{G_{\varphi}} \right\|_{\infty} \|f\|.
\end{aligned} \quad (12)$$

Since P is an orthogonal projection, we have $\|P\| = 1$. Therefore T is a contraction if the maximal gap δ is small enough such that

$$\lambda \delta \left\| \frac{G_{\varphi'}}{G_{\varphi}} \right\|_{\infty} < 1. \quad (13)$$

This is formulated in the following theorem.

Theorem 1: Assume that $\varphi \in W^1$ is a differentiable generator. Then any $f \in V_{\lambda}(\varphi)$ can be reconstructed from any δ -dense set X by the following iterative algorithm:

$$\begin{cases} f_{n+1} = PS_X(f) + (I - PS_X)(f_n) \\ f_0 = PS_X(f), \end{cases} \quad (14)$$

provided that

$$\delta < \left\| \frac{G_{\varphi}}{\lambda G_{\varphi'}} \right\|_0. \quad (15)$$

In practical application, one needs to know the explicit expression for P and S_X . Define ψ by $\hat{\psi} = \hat{\varphi}/G_{\varphi}^2$. Then for any $f \in L^2(\mathbb{R})$, one has

$$P(f) = \sum_k \langle f, \psi(\cdot - k) \rangle \varphi(\cdot - k).$$

Define the *Voronoi domain* V_k of the sampling points t_k as

$$V_k = \{t: |t_k - t| < |t_j - t|, k \neq j\}. \quad (16)$$

Let χ_{V_k} be the characteristic function of V_k . Then for any sampling set $X = \{t_k\}_k$, one can choose a simple interpolation function $S_X(f)$ as

$$S_X(f) = \sum_k f(t_k) \chi_{V_k}. \quad (17)$$

In this special case, one can improve the estimate (15) by a factor π by using Wirtinger's inequality as in [31], that is, (15) becomes

$$\delta < \left\| \frac{\pi G_{\varphi}}{\lambda G_{\varphi'}} \right\|_0. \quad (18)$$

Let $c = \lambda \delta \|G_{\varphi'} G_{\varphi}^{-1}\|_{\infty}$. Then $c < 1$ if δ satisfies (15). From (12), we have

$$(1 - c)\|f\| \leq \|S_X(f)\| \leq (1 + c)\|f\|. \quad (19)$$

Moreover, if X is separated with separation d , then

$$\|S_X(f)\|^2 = \left\| \sum_k f(t_k) \chi_{X_k} \right\|^2 \geq d \sum_k |f(t_k)|^2. \quad (20)$$

On the other hand, since X is δ -dense, we have

$$\|S_X(f)\|^2 = \left\| \sum_k f(t_k) \chi_{X_k} \right\|^2 \leq \delta \sum_k |f(t_k)|^2. \quad (21)$$

By (19), (20) and (21), we derive

$$\frac{1 - c}{\sqrt{\delta}} \|f\| \leq \left(\sum_k |f(t_k)|^2 \right)^{1/2} \leq \frac{1 + c}{\sqrt{d}} \|f\|. \quad (22)$$

This is summarized in the following corollary.

Corollary 1: In Theorem 1, the X is a stable sampling set for $V_{\lambda}(\varphi)$ if X is separated.

When X is a stable sampling set for $V_{\lambda}(\varphi)$, an explicit reconstruction formula is found by us in [10].

If B satisfies the *Strang-Fix condition* [35], that is $\hat{B}(2k\pi) = \delta(k)$. Then $\sum_k B(\cdot - k) = 1$ by the *Poisson Summation Formula*. For a uniform sampling set $X = \{\tau k\}_k$, if B is supported on $[-D, D]$, then we can define S_X by

$$S_X(f) = \sum_k f(\tau k) B(\cdot/\tau - k).$$

Then Theorem 1 holds if $\tau < \|(G_{\varphi}/\lambda D G_{\varphi'})\|_0$.

III. CONCLUSION AND EXAMPLES

Prefiltering a signal by a shift invariant space $V_{\lambda}(\varphi)$ provides more flexible choices to design an A/D conversion system of high accuracy. This paper focuses on the nonuniform sampling in a shift invariant space $V_{\lambda}(\varphi)$. We obtain a formula to calculate the maximal gap of a sampling set for the iterative reconstruction algorithm in the shift invariant space $V_{\lambda}(\varphi)$ in terms of the generator φ , that is

$$\delta < \left\| \frac{G_{\varphi}}{\lambda G_{\varphi'}} \right\|_0. \quad (23)$$

When the interpolation function is chosen to be a simple function, one can improve (23) by a factor of π and obtain the improved estimate

$$\delta < \left\| \frac{\pi G_{\varphi}}{\lambda G_{\varphi'}} \right\|_0. \quad (24)$$

To construct a practical A/D conversion in a shift invariant spaces, it is necessary to understand the maximal gap of sampling set for performing sampling in a shift invariant space. Let's look at some practical examples.

Example 1: Meyer scaling function φ is defined by

$$\hat{\varphi} = \begin{cases} 1, & |\omega| \leq \frac{2\pi}{3}, \\ \cos \left[\frac{\pi}{2} v \left(\frac{3}{2v} |\omega| - 1 \right) \right], & \frac{2\pi}{3} \leq |\omega| \leq \frac{4\pi}{3}, \\ 0, & \text{otherwise} \end{cases}$$

where $v \in C^\infty$, $v(\omega) = 1$ for $\omega \geq 1$, $v(\omega) = 0$ for $\omega \leq 0$, and $v(\omega) + v(1 - \omega) = 1$. Then $G_{\varphi} = 1$. Since $G_{\varphi'} \leq 4\pi/3$, we obtain the maximal gap

$$\delta < \frac{3}{4\pi\lambda}.$$

Example 2: B-spline β^N of degree N is defined by the N times convolution of the characteristic function of the interval $[0, 1]$, i.e., $\beta^N = \chi_{[0,1]} * \dots * \chi_{[0,1]}$. Then $\|G_{\varphi}\|_0 \geq ((2/\pi))^{N+1}$ and $\|G_{\varphi'}\|_\infty \leq \sqrt{(2N+1)/(2N-1)}$. Therefore, the maximal gap is

$$\delta < \sqrt{\frac{2N-1}{2N+1}} \left(\frac{2}{\pi} \right)^{N+1}.$$

Example 3: Gaussian kernel K is defined by $K(t) = 1/(\sqrt{2\pi})e^{-t^2/2}$. Then $\hat{K}(\omega) = e^{-\omega^2/2}$, and we have $\|G_{\varphi}\|_0 \geq e^{-\pi^2}$ and $\|G_{\varphi'}\|_\infty \leq 1 + 2e^{-\pi^2}$. Therefore, we obtain the maximal gap

$$\delta < \frac{e^{-\pi^2}}{\lambda(1 + 2e^{-\pi^2})}.$$

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