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Recovery of Missing Samples in Multi-channel Oversampling of Multi-banded Signals

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Abstract**—** We show that in a two-channel sampling series expansion of band-pass signals, any finitely many missing samples can always be recovered via oversampling in a larger band-pass region. We also obtain an analogous result for multi-channel oversampling of harmonic signals.

Keywords**—** oversampling, multi-channel sampling, recovery of missing samples, band-pass signal, harmonic signal

I. INTRODUCTION

FOR a bounded and closed band-region B, let PW_B be the Paley-Wiener space of finite energy (i.e. square integrable) signals of which frequencies are confined in B. That is,

$$PW_B := \{f(t) \in L^2(\mathbb{R}) : \text{supp } \hat{f}(\xi) \subset B\},$$

where $\mathcal{F}(f)(\xi) = \hat{f}(\xi) := \frac{1}{\sqrt{2\pi}}\int_{-\infty}^{\infty} f(t)e^{-it\xi}dt$ is the Fourier transform of $f(t)$ with inverse Fourier transform $f(t) = \mathcal{F}^{-1}(\hat{f})(t) := \frac{1}{\sqrt{2\pi}}\int_{-\infty}^{\infty}\hat{f}(\xi)e^{it\xi}d\xi$.
If a signal $f(t)$ is single-banded with band-region $B = [-\pi\omega, \pi\omega]$ $(\omega > 0)$, then $f(t)$ can be expanded as a Shannon sampling series:

$$f(t) = \sum_{n\in\mathbb{Z}} f\Big(\frac{n}{\omega}\Big)\frac{\sin \pi(t-n)}{\pi(t-n)} \ ,$$

in which all samples $\{f(\frac{n}{\omega}) : n \in \mathbb{Z}\}$ are independent. However, if we oversample $f(t)$ with higher rate than the optimal Nyquist rate ω, then the resulting samples are dependent. Using this observation, we may recover finitely many missing samples([2,3,5,8]). When we join oversampling and multi-channeling, we may or may not able to recover finitely missing samples depending on the nature of the band-region B and pre-filters used in channeling ([6,10]). In this work, we show that in case of band-pass and harmonic signals, any finitely many missing samples can be always recovered through a multi-channel oversampling in a larger band-region of the same type.

II. OVERSAMPLING OF BAND-PASS SIGNALS

Consider a band-pass region $B = B_- \cup B_+$, where $w_0, w > 0$ and

$$B_- = [-\pi(\omega_0+\omega), -\pi\omega_0] \text{ and } B_+ = [\pi\omega_0, \pi(\omega_0+\omega)].$$

Then the optimal Nyquist rate for signals in PW_B is ω samples per second. For τ with $0 < \tau \le w_0$, let $\tilde{B} = \tilde{B}_- \cup \tilde{B}_+$ be another band-pass region, where

$$\tilde{B}_- = [-\pi(\omega_0+\omega+\tau), -\pi(\omega_0-\tau)]$$

and

$$\tilde{B}_+ = [\pi(\omega_0-\tau), \pi(\omega_0+\omega+\tau)].$$

We take τ so that $r := \frac{2\omega_0+\omega}{2\tau+\omega}$ is a positive integer. Then $\tilde{B}_+ = \tilde{B}_- + r\pi(2\tau+\omega)$ so that $\tilde{B}$ becomes a so-called selectively tiled band-region([4]) of length $2\pi\tilde{\omega}$ with $\tilde{\omega} = \omega + 2\tau$. Note that the smallest such τ is obtained when we take r to be the largest integer less than $1 + \frac{2\omega_0}{\omega}$. We now take two pre-filters of bounded measurable functions $A_j(\xi)$ $(j = 1, 2)$ on $\tilde{B}$. We set

$$A(\xi) = \begin{bmatrix} A_1(\xi) & A_1(\xi + r\pi\tilde{\omega}) \\ A_2(\xi) & A_2(\xi + r\pi\tilde{\omega}) \end{bmatrix} \text{ on } \tilde{B}_-$$

and assume for some constant $\alpha > 0$, $|\det A(\xi)| \ge \alpha$ a.e. on $\tilde{B}_-$.

For any band-pass signal $f(t)$ in $PW_{\tilde{B}}$, let

$$c_j(f)(t) := \mathcal{F}^{-1}(A_j(\xi)\hat{f}(\xi))(t) = \frac{1}{\sqrt{2\pi}}\int_{\tilde{B}} A_j(\xi)\hat{f}(\xi)e^{it\xi}d\xi \tag{1}$$

be the channeled output signals of the input signal $f(t)$. Then ([4,7,8,9])

$$f(t) = \sum_{j=1}^{2}\sum_n c_j(f)\Big(\frac{2n}{\tilde{\omega}}\Big)S_{j,n}(t), \tag{2}$$

which converges in $PW_{\tilde{B}}$ and also converges uniformly on $\mathbb{R}$. By taking Fourier transform on (2), we obtain

$$\hat{f}(\xi) = \sum_{j=1}^{2}\sum_n c_j(f)\Big(\frac{2n}{\tilde{\omega}}\Big)\phi_{j,n}(\xi),$$

which converges in $L^2(B)$, where

$$\phi_{j,n}(\xi) = \frac{1}{\tilde{\omega}}\sqrt{\frac{2}{\pi}}U_j(\xi)e^{-i\frac{2n}{\tilde{\omega}}\xi}$$

and

$$A(\xi)^{-1} = \begin{bmatrix} U_1(\xi) & U_2(\xi) \\ U_1(\xi + r\pi\tilde{\omega}) & U_2(\xi + r\pi\tilde{\omega}) \end{bmatrix} \text{ on } \tilde{B}_-. \tag{3}$$

If $f(t)$ is in PW_B, i.e., supp $\hat{f} \subset B$, then

$$\hat{f}(\xi) = \sum_{j=1}^{2}\sum_n c_j(f)\Big(\frac{2n}{\tilde{\omega}}\Big)\phi_{j,n}(\xi)\chi_B(\xi) \text{ in } L^2(B), \tag{4}$$

where $\chi_B(\xi)$ is the characteristic function of B. By taking inverse Fourier transform on (4), we have

$$f(t) = \sum_{j=1}^{2}\sum_n c_j(f)\Big(\frac{2n}{\tilde{\omega}}\Big)T_{j,n}(t) \tag{5}$$

where $T_{j,n}(t) = \frac{1}{\sqrt{2\pi}}\int_B \phi_{j,n}e^{it\xi}d\xi$. We may call (5) a two-channel oversampling series expansion of $f(t)$ in PW_B.

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III. Recovering missing samples

For a band-pass signal $f(t)$ in PW_B, consider its oversampled expansion (5).

Lemma 1. *We have for any integer* m

$$c_k(f)\Big(\frac{2m}{\tilde{\omega}}\Big) = \frac{1}{\pi\tilde{\omega}}\sum_n c_k(f)\Big(\frac{2n}{\tilde{\omega}}\Big)\int_{B_-} e^{i\frac{2}{\tilde{\omega}}(m-n)\xi} \quad (6)$$

for $k = 1, 2$.

Proof: By (1) and (4), we have

$$c_k(f)(t) = \frac{1}{\sqrt{2\pi}}\int_{\tilde{B}} A_k(\xi)\hat{f}(\xi)e^{it\xi}d\xi$$
$$= \frac{1}{\pi\tilde{\omega}}\sum_{j=1}^{2}\sum_n c_j\Big(\frac{2n}{\tilde{\omega}}\Big)\int_{\tilde{B}} A_k(\xi)U_j(\xi)\chi_B(\xi)e^{i(t-\frac{2n}{\tilde{\omega}})\xi}d\xi.$$

Hence for any integer m we have

$$c_k(f)\Big(\frac{2m}{\tilde{\omega}}\Big) =$$
$$\frac{1}{\pi\tilde{\omega}}\sum_{j=1}^{2}\sum_n c_j\Big(\frac{2n}{\tilde{\omega}}\Big)\Big[\int_{\tilde{B}_-} A_k(\xi)U_j(\xi)\chi_B(\xi)e^{i\frac{2}{\tilde{\omega}}(m-n)}d\xi$$
$$+\int_{\tilde{B}_+} A_k(\xi)U_j(\xi)\chi_B(\xi)e^{i\frac{2}{\tilde{\omega}}(m-n)}d\xi\Big]$$
$$= \frac{1}{\pi\tilde{\omega}}\sum_{j=1}^{2}\sum_n c_j\Big(\frac{2n}{\tilde{\omega}}\Big)\int_{\tilde{B}_-}\Big[A_k(\xi)U_j(\xi)$$
$$+ A_k(\xi + r\pi\tilde{\omega})U_j(\xi + r\pi\tilde{\omega})\Big]\chi_{B_-}(\xi)e^{i\frac{2}{\tilde{\omega}}(m-n)}d\xi,$$

from which (6) comes since $A_k(\xi)U_j(\xi) + A_k(\xi+\pi\tilde{\omega})U_j(\xi+\pi\tilde{\omega}) = \delta_{jk}$ by (3). □

Theorem 1. *For any finite index sets of integers* I_1 *and* I_2*, any finite missing samples* $\{c_1(f)(\frac{2m}{\tilde{\omega}}) : m \in I_1\} \cup \{c_2(f)(\frac{2n}{\tilde{\omega}}) : n \in I_2\}$ *can be uniquely recovered.*

Proof: Set $I_1 = \{m_1, m_2, \cdots, m_M\}$ if $I_1 \neq \phi$ and $I_2 = \{n_1, n_2, \cdots, n_N\}$ if $I_2 \neq \phi$. Then we have from (6)

$$c_1(f)\Big(\frac{2m_j}{\tilde{\omega}}\Big) = \frac{1}{\pi\tilde{\omega}}\sum_{k=1}^{M} r(m_j, m_k)c_1(f)\Big(\frac{2m_k}{\tilde{\omega}}\Big) + g_{1j} \quad (7)$$

for $1 \le j \le M$ and

$$c_2(f)\Big(\frac{2n_j}{\tilde{\omega}}\Big) = \frac{1}{\pi\tilde{\omega}}\sum_{k=1}^{N} r(n_j, n_k)c_2(f)\Big(\frac{2n_k}{\tilde{\omega}}\Big) + g_{2j} \quad (8)$$

for $1 \le j \le N$ where g_{1j}'s and g_{2j}'s are known quantities and

$$r(m,n) := \int_{B_-} e^{i\frac{2}{\tilde{\omega}}(m-n)\xi}d\xi \text{ for } m,n \in \mathbb{Z}.$$

We may write (7-8) in a vector form as :

$$\begin{cases} (I - S_1)\mathbf{c}_1 = \mathbf{g}_1 \\ (I - S_2)\mathbf{c}_2 = \mathbf{g}_2 \end{cases} \quad (9)$$

where

$$\mathbf{c}_1 := \Big(c_1(f)\Big(\frac{2m_1}{\tilde{\omega}}\Big), \cdots, c_1(f)\Big(\frac{2m_M}{\tilde{\omega}}\Big)\Big)^T,$$
$$\mathbf{c}_2 := \Big(c_2(f)\Big(\frac{2n_1}{\tilde{\omega}}\Big), \cdots, c_2(f)\Big(\frac{2n_N}{\tilde{\omega}}\Big)\Big)^T,$$
$$\mathbf{g}_1 := (g_{11}, \cdots, g_{1M})^T,$$
$$\mathbf{g}_2 := (g_{21}, \cdots, g_{2N})^T,$$

and

$$S_1 = \Big[\frac{1}{\pi\tilde{\omega}}r(m_j, m_k)\Big]_{j,k=1}^{M},\ S_2 = \Big[\frac{1}{\pi\tilde{\omega}}r(n_j, n_k)\Big]_{j,k=1}^{N}.$$

Note that S_1 and S_2 are self-adjoint. Now for any $u = (u_1, \cdots, u_M) \in \mathbb{C} \setminus \{0\}$,

$$\langle S_1 u, u\rangle = \frac{1}{\pi\tilde{\omega}}\sum_{j,k=1}^{M} r(m_j, m_k)u_k\overline{u_j}$$
$$= \int_{\tilde{B}_-}\Big|\sum_{j=1}^{M}\overline{u_j}\frac{1}{\sqrt{\pi\tilde{\omega}}}e^{i\frac{2}{\tilde{\omega}}m_j\xi}\Big|^2\chi_{B_-}(\xi)d\xi$$
$$< \int_{\tilde{B}_-}\Big|\sum_{j=1}^{M}\overline{u_j}\frac{1}{\sqrt{\pi\tilde{\omega}}}e^{i\frac{2}{\tilde{\omega}}m_j\xi}\Big|^2 d\xi = \sum_{j=1}^{M}|u_j|^2$$

since $\{\frac{1}{\sqrt{\pi\tilde{\omega}}}e^{i\frac{2}{\tilde{\omega}}m\xi}\}_{m\in\mathbb{Z}}$ is an orthonormal basis of $L^2(\tilde{B}_-)$. Hence, 1 cannot be an eigenvalue of S_1. Similarly, 1 cannot be an eigenvalue of S_2. Therefore, both equations in (9) have unique solutions $\mathbf{c}_1$ and $\mathbf{c}_2$. □

Above process can be readily extended to multi-channel oversampling of harmonic signals (see [1] and Chaper 13 in [4]). Let $f(t)$ be a harmonic signal in PW_B, where

$$B := \bigcup_{i=1}^{N}[a_i, b_i]$$

is a harmonic band-region and

$$b_i - a_i = \pi\omega\ (1 \le i \le N)$$
$$a_{i+1} - b_i = 2\pi\omega_0\ (1 \le i < N) \text{ for } \omega,\ \omega_0 > 0.$$

For $0 < \tau \le \omega_0$, let $\tilde{B} := \cup_{i=1}^{N}\tilde{B}_i$ be another harmonic band-region, where

$$\tilde{B}_i = [a_i - \pi\tau, b_i + \pi\tau] \text{ for } 1 \le i \le N.$$

We take τ so that $r := \frac{2\omega_0+\omega}{2\tau+\omega}$ is a positive integer. Then $\tilde{B}_j = \tilde{B}_i + (j-i)r\pi(2\tau+\omega)$ for $1 \le i < j \le N$ so that $\tilde{B}$ becomes a so-called selectively tiled band-region of total length $N\pi\tilde{\omega}$, where $\tilde{\omega} = \omega + 2\tau$. We now take N pre-filters $A_j(\xi)$ $(j = 1, 2, \cdots, N)$ of bounded measurable functions on $\tilde{B}$. We set $A(\xi)$ be the $N \times N$ matrix whose (j,k)th component is given by

$$A_{jk}(\xi) = A_j(\xi + (k-1)r\pi\tilde{\omega})$$

and assume $|\det A(\xi)| \ge \alpha > 0$ a.e. on $\tilde{B}_1$. Let

$$c_j(f)(t) := \mathcal{F}^{-1}(A_j(\xi)\hat{f}(\xi))(t) = \frac{1}{\sqrt{2\pi}}\int_{\tilde{B}} A_j(\xi)\hat{f}(\xi)e^{it\xi}d\xi$$

be the channeled output signals. Proceeding as in Section 2, we can obtain an oversampling formula for any harmonic signal $f(t)$ in PW_B (but viewed as a signal in $PW_{\tilde{B}}$) as

$$f(t) = \sum_{j=1}^{N} \sum_{n} c_j(f)\left(\frac{2n}{\tilde{\omega}}\right) T_{j,n}(t). \tag{10}$$

Then, we have the following multi-channel analog of Theorem 3.2.

Theorem 2. *For any finite index sets of integers* $I_i (i = 1, 2, \cdots, N)$*, any finite missing samples* $\cup_{i=1}^{N} \{c_i(f)(\frac{2m}{\tilde{\omega}}) : m \in I_i\}$ *from the oversampling (10) can be uniquely recovered.*

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