Imaging sampling below the Nyquist density without aliasing

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For multidimensional band-limited functions, the Nyquist density is defined as that density corresponding to maximally packed spectral replications. Because of the shape of the support of the spectrum, however, sampling multidimensional functions at Nyquist densities can leave gaps among these replications. In this paper we show that, when such gaps exist, the image samples can periodically be deleted or decimated without information loss. The result is an overall lower sampling density. Recovery of the decimated samples by the remaining samples is a linear interpolation process. The interpolation kernels can generally be obtained in closed form. The interpolation noise level resulting from noisy data is related to the decimation geometry. The greater the clustering of the decimated samples, the higher the interpolation noise level is.

INTRODUCTION

For a band-limited function of any dimension, the Nyquist density is defined as the periodic sampling density that results in the most densely packed replications of the function's spectrum.¹ Thus, for one-dimensional (1-D) functions band limited in the low-pass sense, the Nyquist density (or rate) is equal to the extent of the support of the function's spectrum. The samples' spectrum is maximally packed at this rate and there are no gaps among the spectral replications. Because of the more complicated geometry of the spectrum's support, this is not generally true for functions of higher dimension. The extent of the support is measured not only by the area but also by the shape. Consider, for example, a two-dimensional (2-D) circularly bandlimited function. The Nyquist density is the density corresponding to a hexagonal replication pattern (Fig. 1).² Clearly, gaps exists among the replications. We will show that, if gaps exists among the replications, then the samples are linearly dependent. If the samples are linearly dependent, the function is said to be oversampled. Hence a multidimensional band-limited function can be oversampled when it is sampled at its Nyquist density.

Marks¹ showed that, if gaps exist among the spectral replications, a subset of samples can be deleted and restored by a linear combination of those remaining. In this paper we extend this result to deletion of an infinite number of samples. Specifically, given a sample set whose spectrum contains gaps, we show that samples can periodically be deleted or decimated. The sampling density, equal to the original uniform density minus the deletion density, is therefore reduced. If the original sample set is obtained by sampling the function at its Nyquist density, decimation will reduce the density to below that of Nyquist. The deleted samples can be restored by a linear interpolation of those samples remaining. The required interpolation kernels can be evaluated in closed form. The noise sensitivity of this process is also investigated.

PRELIMINARY DESCRIPTIONS

Before we proceed to the contribution of this paper, a brief review of the multidimensional sampling theorem and related work is in order.

Multidimensional Sampling Theorem

The multidimensional sampling theorem, initially presented by Petersen and Middleton,³ is a direct extension of the Shannon sampling theorem.⁴ Let $\{x(t)|t = (t_1, t_2, \ldots, t_N)^T\}$ be an N-dimensional signal (the subscript T denotes transposition). The Fourier transform of x(t) is

$$X(\mathbf{\nu}) = \int_{\mathbf{t}} x(\mathbf{t}) \exp(-j2\pi \mathbf{\nu}^T \mathbf{t}) d\mathbf{t}, \qquad (1)$$

where

and

$$\int_{\mathbf{t}} \mathbf{dt} = \int_{t_1} \mathbf{d}t_1 \int_{t_2} \mathbf{d}t_2 \dots \int_{t_N} \mathbf{d}t_N$$

 $\boldsymbol{\nu} = (\nu_1, \nu_2, \ldots, \nu_N)^T$

If X(v) is zero outside a support region, \mathcal{A} , contained within a hypersphere of finite radius, we say that x(t) is \mathcal{A} band limited. The inverse Fourier transformation follows as

$$\mathbf{x}(\mathbf{t}) = \int_{\mathbf{v}\in\mathcal{A}} X(\mathbf{v}) \exp(j2\pi\mathbf{v}^T \mathbf{t}) d\mathbf{v}.$$
 (2)

When x(t) is sampled with a regular sampling geometry, the spectrum of the sampled signal replicates periodically. Let $\{v_n | 1 \le n \le N\}$ be the N sampling vectors, and let k_s be an offset vector. These N + 1 vectors constitute the sampling geometry for sampling x(t). In particular, the sampling geometry is denoted by the matrix-vector pair $[V, k_s]$. Let $\hat{x}(t)$ be the signal containing the sample impulses,



Fig. 1. (a) Hexagonal sampling lattice corresponding to the Nyquist-density sampling of a circularly band-limited function, (b) resulting spectral replications.

$$\hat{x}(\mathbf{t}) = \sum_{\mathbf{n}} x(\mathbf{t} + \mathbf{V}\mathbf{k}_{s})\delta(\mathbf{t} - \mathbf{V}\mathbf{n})$$
$$= \sum_{\mathbf{n}} x[\mathbf{V}(\mathbf{n} + \mathbf{k}_{s})]\delta(\mathbf{t} - \mathbf{V}\mathbf{n}), \qquad (3)$$

where

$$\mathbf{V} = [\mathbf{v}_1 | \mathbf{v}_2 | \dots | \mathbf{v}_N]$$

is the sampling matrix, $\mathbf{n} = (n_1, n_2, \dots, n_N)^T$ is a vector of integers, $\delta(\mathbf{t})$ is an N-dimensional Dirac delta function, and

$$\sum_{\mathbf{n}} = \sum_{n_1} \sum_{n_2} \dots \sum_{n_N}.$$

The sampling matrix VM gives the same sampling geometry

as V when the matrix of integers, \mathbf{M} , has a determinant magnitude of 1.

With no loss of generality, we let \mathbf{k}_s be the zero vector. The replicated spectrum of the samples follows straightforwardly from the Poisson sum formula:

$$\hat{X}(\boldsymbol{\nu}) = D \sum_{\mathbf{n}} X(\boldsymbol{\nu} - \mathbf{U}\mathbf{n}), \qquad (4)$$

where U is the periodicity matrix,

$$\mathbf{U} = [\mathbf{u}_1 | \mathbf{u}_2 | \dots | \mathbf{u}_N],$$

and D is the sampling density,

$$D = |\mathbf{U}| = \frac{1}{|\mathbf{V}|} \quad \text{(samples per area)}, \tag{5}$$



Fig. 2. (a) Example of a sampling lattice and (b) its corresponding spectral replications. Each dashed-line parallelogram is a cell \mathcal{C} .



Fig. 3. Two different cells corresponding to the same periodicity matrix. In this example, the puzzle pieces A, B, C, and D are simply rearranged.

and we have adopted the notation $|\mathbf{U}|$ to denote the determinant of the matrix \mathbf{U} . The N column vectors of \mathbf{U} , referred to as the replication vectors, are obtained from

$$\mathbf{U}^T = \mathbf{V}^{-1}.\tag{6}$$

Clearly, we require that V be full rank or, equivalently, that the N sampling vectors be linearly independent.

An example of a 2-D sampling geometry is shown in Fig. 2. A period of the samples' spectrum, $\hat{X}(\nu)$, denoted \mathcal{C} , is a geometric body measured not only by the extent (or area) but also by the shape. For a given sampling matrix V, the area of the corresponding \mathcal{C} is always equal to the sampling density D, although, as illustrated in Fig. 3, there is more than one possible shape. A parallelepiped is an obvious shape choice because of its ease of construction. Each periodicity vector, \mathbf{u}_n , is one of the legs. We will refer to a spectral period as a cell. Different methods for constructing \mathcal{C} from U were discussed by Dubois.⁵ The area of a cell is clearly equal to $|\mathbf{U}|$, which, from Eq. (5), is equal to the sampling density D.

If there is no aliasing in $\hat{X}(\nu)$, we can recover the original signal, x(t), by passing $\hat{x}(t)$ through an ideal low-pass filter of

height 1/D that passes only the zeroth-order spectrum [the term corresponding to $\mathbf{n} = \mathbf{0}$ in Eq. (4)]. The resulting cardinal series is

$$x(\mathbf{t}) = \sum_{\mathbf{n}} x(\mathbf{V}\mathbf{n}) f(\mathbf{t} - \mathbf{V}\mathbf{n}), \tag{7}$$

where the interpolation kernel is obtained by

$$f(\mathbf{t}) = \frac{1}{D} \int_{\mathcal{B}} \exp(-j2\pi\nu^{T} \mathbf{t}) \mathrm{d}\nu$$
(8)

and \mathcal{B} is any region enclosing only the zeroth-order spectrum. For example, \mathcal{B} could be over \mathcal{C} or the support \mathcal{A} (Fig. 4).

For all 1-D low-pass band-limited functions, gaps cease to exist when the function is sampled at the Nyquist rate. This, however, is not generally true in higher dimensions. In our running example, because of the circular shape of the support, gaps still exist when the function is sampled at the Nyquist density. In the following subsection we discuss the relationship between sample interdependency and gaps in the samples' spectrum.

Sample Interdependency

If gaps exist among the spectral replications, then the samples $\{x(\mathbf{Vn})\}\$ are linearly dependent. This relationship can easily be demonstrated in cases of 1-D band-limited signals. Let x(t) be a 1-D signal whose spectrum, $X(\nu)$, is identically zero only for $|\nu| \ge \lambda$. To sample x(t) at above the Nyquist rate corresponds to a sampling frequency of $2S > 2\lambda$. The samples' spectrum $\hat{X}(\nu)$ is periodic replication of $X(\nu)$ with period 2S,

$$\hat{X}(\nu) = 2S \sum_{n=-\infty}^{\infty} X(\nu - 2nS).$$

Since $S > \lambda$, there exists a periodic disjoint region, \mathcal{D} , where $\hat{X}(\nu)$ is identically zero (Fig. 5):

$$\hat{X}(v) \equiv 0, \quad v \in \mathcal{D}.$$



Fig. 4. Region of integration, \mathcal{B} , of the interpolation kernel in Eq. (8).



Fig. 5. Example of a spectral replication caused by the oversampling of a function.

Let $H(\nu)$ be the impulse response of some bandpass filter whose magnitude is nonzero over \mathcal{D} only. Then

$$H(\nu)\hat{X}(\nu) \equiv 0, \tag{9}$$

or, for every t,

,

$$\sum_{n=-\infty}^{\infty} x\left(\frac{n}{2S}\right) h\left(t - \frac{n}{2S}\right) \equiv 0.$$
 (10)

Since h(t) is nowhere identically zero over a nonzero interval, we can view Eq. (10) as a linear combination of the samples with nonzero coefficients. Clearly, the samples are linearly interdependent. The linear dependence among samples ceases when x(t) is sampled at the Nyquist rate ($S = \lambda$). At this rate, the gap region vanishes, and therefore h(t) is identically zero for every t.

As shown in Fig. 1, gaps can exist among the spectral replications in higher dimensions. Hence the samples are linearly dependent even if the signal is sampled at the Nyquist density. A signal is oversampled if the samples are linearly dependent. Thus multidimensional band-limited signals may be oversampled even if they are sampled at Nyquist densities. Exceptions are cases in which the support is a cell (i.e., $\mathcal{A} = \mathcal{O}$).

Sample Deletion and Restoration

Given a sample set that is linearly dependent, one can reduce dependency by deleting some samples. Indeed, Marks showed that if a signal is oversampled, a finite subset of samples can be deleted and restored as a linear combination of those remaining.

Let \mathcal{M} denote the set of indices of the M deleted samples located at the points $\{\mathbf{Vm} | \mathbf{m} \in \mathcal{M}\}$. We can thus rewrite Eq. (10) as follows:

$$\sum_{\mathbf{m}\in\mathcal{M}} x(\mathbf{V}\mathbf{m})h(\mathbf{t}-\mathbf{V}\mathbf{m}) = -\sum_{\mathbf{n}\notin\mathcal{M}} x(\mathbf{V}\mathbf{n})h(\mathbf{t}-\mathbf{V}\mathbf{n}).$$
(11)

By evaluating Eq. (11) at the M locations of the deleted samples, we obtain a system of M equations:

$$\sum_{\mathbf{m}\in\mathcal{M}} x(\mathbf{V}\mathbf{m})h[\mathbf{V}(\mathbf{l}-\mathbf{m})] = -\sum_{\mathbf{n}\notin\mathcal{M}} x(\mathbf{V}\mathbf{n})h[\mathbf{V}(\mathbf{l}-\mathbf{m})],$$
$$\mathbf{l}\in\mathcal{M}. (12)$$

This system of equations can be put into a matrix form:

$$\mathbf{H}_{\mathcal{M}}\mathbf{x}_{\mathcal{M}} = -\mathbf{H}_{R}\mathbf{x}_{R},\tag{13}$$

where $\mathbf{x}_{\mathcal{M}}$ is an *M*-dimensional vector whose entries are the deleted samples, \mathbf{x}_R is a vector whose entries are the remaining samples, and $\mathbf{H}_{\mathcal{M}}$ is a square matrix of dimension *M*. One can show that $\mathbf{H}_{\mathcal{M}}$ is always invertible.⁶ The deleted samples can thus be restored by a linear combination of the remaining samples:

$$\mathbf{x}_{\mathcal{M}} = -\mathbf{H}_{\mathcal{M}}^{-1}\mathbf{H}_{R}\mathbf{x}_{R}.$$
 (14)

The quality of restoration of the deleted samples depends on the condition of $\mathbf{H}_{\mathcal{M}}$, which, in turn depends on three factors: (1) the distance among the deleted samples, (2) the uniformity of the deletion geometry,^{7,8} and (3) the total number of the deleted samples. Ching⁹ performed an extensive empirical examination of the effect of these three factors on the quality of sample restoration.

SAMPLING DECIMATION (PERIODIC DELETION)

In this section we present a second deletion scheme wherein samples are periodically deleted or decimated. Since the samples are deleted periodically, the overall sampling density is reduced. We will show that, as long as gaps exist among the spectral replications, samples can be thus decimated.

Decimation Lattice

Using the notation of sampling geometry, we denote the decimation geometry by $[\mathbf{V}_d, \mathbf{k}_d]$. The vector \mathbf{k}_d is an integer vector that determines the offset of the decimation lattice from the origin. In particular, the decimation matrix, \mathbf{V}_d , is an integer-weighted combination of the sampling vectors. The decimation matrix can therefore be written as

$$\mathbf{V}_{d} = \mathbf{V}\mathbf{M}$$
$$= [\mathbf{v}_{d_{1}}|\mathbf{v}_{d_{2}}|\dots|\mathbf{v}_{d_{N}}], \qquad (15)$$

where the v_{d_i} 's are the N decimation vectors and **M** is an $N \times N$ nonsingular matrix of integers. The location of the deleted samples is the set of lattice points $\{V_d \mathbf{n} + V \mathbf{k}_d\} = \{V(\mathbf{Mn} + \mathbf{k}_d)\}.$

With reference to Eq. (5), the decimation density is equal to

$$D_d = \frac{1}{|\mathbf{V}_d|} = \frac{D}{|\mathbf{M}|},\tag{16}$$

where D is the sampling density. The decimation ratio, r_d , defined as the fraction of the samples deleted, is the ratio of D_d to D:



Fig. 6. (a) Example of a decimation geometry imposed upon the samples in Fig. 2, (b) corresponding subcell pattern (dashed-line squares).

$$r_d = \frac{1}{|\mathbf{M}|}.\tag{17}$$

As an example, the lattice defined in Fig. 2 is redrawn in Fig. 6. The open circles form the deletion lattice with

$$\mathbf{M} = \begin{bmatrix} 2 & 0 \\ 1 & 2 \end{bmatrix}$$

and

$$\mathbf{k}_d = (1, 1)^T$$
.

A parallelepiped constructed by the N decimation vectors forms a decimation period. Let L be the number of samples enclosed within a decimation period. A decimation period, corresponding to our running example, is outlined in a dashed-line parallelogram in Fig. 6(a), in which the period contains L = 4 points. Obviously,

$$L = |\mathbf{M}|. \tag{18}$$

Let L_j be the length of the *j*th side of the parallelepipedshaped decimation period. Then

$$L=\prod_{j=1}^N L_j$$

To produce a structure analogous to the polyphase structure in multirate digital signal processing,¹⁰ we divide the sample set $\{x(\mathbf{Vn})\}$ into L subgroups:

$$x_i(\mathbf{V}_d\mathbf{n}) = x(\mathbf{V}_d\mathbf{n} + \mathbf{V}\mathbf{k}_i), \qquad i = 0 \text{ to } L - 1, \tag{19}$$

where \mathbf{k}_i is an integer vector (which is also a mixed-radix representation of the integer *i*). The radix of the *j*th element of \mathbf{k}_i is L_j . The original sample set is a union of the L sample subgroups,

$$\{\mathbf{x}(\mathbf{Vn})\} = \bigcup_{i=0}^{L-1} \{\mathbf{x}_i(\mathbf{V}_d\mathbf{n} - \mathbf{Vk}_i)\}.$$
 (20)

Let $\hat{X}_i(\nu)$ be the spectrum of the *i*th sample subgroup; then, from Eq. (19),

$$\hat{X}(\boldsymbol{\nu}) = \sum_{i=0}^{L-1} \hat{X}_i(\boldsymbol{\nu}) \exp(-j2\pi\boldsymbol{\nu}^T \mathbf{V} \mathbf{k}_i).$$
(21)

If one of the L subgroups can be deleted, then the deletion is periodic, and the sampling density is reduced by a factor of 1/L.

The periodicity matrix corresponding to V_d is

$$\mathbf{U}_d^{T} = \mathbf{V}_d^{-1}.$$
 (22)

For convenience, we will refer to the cells corresponding to U_d as subcells. Naturally, the area of a subcell, denoted \mathcal{C}_d , is 1/L that of the cell \mathcal{C} . In our running example, square subcells are shown in Fig. 6(b) by dashed lines. Since $|\mathbf{M}| = 4$, the area of the cell \mathcal{C} is 4 times that of the subcell \mathcal{C}_d .

First-Order Decimation and Restoration

If a subcell is totally subsumed in a gap of the samples' spectrum, then one of L subgroups can be deleted by using the decimation geometry $[\mathbf{V}_d, \mathbf{k}_\alpha]$, where α is the index of the deleted subgroup. The deleted samples can be restored with the knowledge of those samples remaining. We can prove this as follows.

Let $\{x_{\alpha}(\mathbf{V}_{d}\mathbf{n})\}$ be the deleted subgroup:

$$x_{\alpha}(\mathbf{V}_{d}\mathbf{n}) = x(\mathbf{V}_{d}\mathbf{n} + \mathbf{V}\mathbf{k}_{\alpha}).$$
(23)

The spectrum of this subgroup is denoted $\hat{X}_{\alpha}(\nu)$. The set of the indices of the decimated samples is thus

$$\mathcal{M} = \{\mathbf{m} | \mathbf{m} = \mathbf{M}\mathbf{n} + \mathbf{k}_{\alpha}\},\tag{24}$$

where n is an arbitrary vector of integers. We rewrite Eq. (21) as

$$\hat{X}(\boldsymbol{\nu}) = \sum_{i \neq \alpha} \hat{X}_i(\boldsymbol{\nu}) \exp(-j2\pi\boldsymbol{\nu}^T \mathbf{V} \mathbf{k}_i) + \hat{X}_\alpha(\boldsymbol{\nu}) \exp(-j2\pi\boldsymbol{\nu}^T \mathbf{V} \mathbf{k}_\alpha).$$
(25)

If there exists a subcell, denoted $\hat{\mathcal{C}}_d$, subsumed within a gap, then

$$\hat{X}(\boldsymbol{\nu}) \equiv 0, \qquad \boldsymbol{\nu} \in \hat{\mathcal{C}}_d. \tag{26}$$

Then, with reference to Eq. (25),

$$\hat{X}_{\alpha}(\boldsymbol{\nu}) = -\sum_{i\neq\alpha} \hat{X}_{i}(\boldsymbol{\nu}) \exp[-j2\pi\boldsymbol{\nu}^{T} \mathbf{V}(\mathbf{k}_{i}-\mathbf{k}_{\alpha})], \quad \boldsymbol{\nu} \in \hat{\mathcal{C}}_{d}.$$
(27)

The spectrum of the deleted subgroup is a linear combination of the other L-1 subgroups. From the theory of the Fourier-series expansion, the decimated samples can be restored with the knowledge of a period of $\hat{X}_{\alpha}(\nu)$ regardless of where the period is located. Hence, with the knowledge of $X_{\alpha}(\nu)$ in $\hat{\mathcal{C}}_d$, the decimated samples can be recovered by

$$\mathbf{x}_{\alpha}(\mathbf{V}_{d}\mathbf{m}) = \frac{1}{D_{d}} \int_{\hat{\mathcal{C}}_{d}} \hat{X}_{\alpha}(\boldsymbol{\nu}) \exp(j2\pi\boldsymbol{\nu}^{T}\mathbf{V}_{d}\mathbf{m}) d\boldsymbol{\nu}.$$
 (28)

By substituting Eq. (27) into Eq. (28), we obtain the interpolation formula to recover the decimated samples:

$$x_{\alpha}(\mathbf{V}_{d}\mathbf{m}) = \sum_{i \neq \alpha} \sum_{\mathbf{n}} x_{i}(\mathbf{V}_{d}\mathbf{n}) f_{d} [\mathbf{V}_{d}(\mathbf{m}-\mathbf{n}) - \mathbf{V}(\mathbf{k}_{i}-\mathbf{k}_{\alpha})],$$
(29)

where

$$f_d(\mathbf{t}) = -\frac{1}{D_d} \int_{\hat{\mathcal{O}}_d} \exp(-j2\pi \boldsymbol{\nu}^T \mathbf{t}) d\boldsymbol{\nu}$$
(30)

is the interpolation kernel.

In summary, if there exists a subcell subsumed in a gap, a subgroup of samples can be decimated. The decimated samples can be restored by those remaining.

Examples

Here we give three examples of first-order decimation for 2-D cases of circular spectral support. A monochromatic coherent or incoherent image of an object of finite extent obtained through an optical imaging system with a circular pupil, for example, has such spectral support.¹¹ In our example, the circle's diameter is set to 1.

Rectangular Sampling

If the sampling geometry is limited to a rectangular lattice, the minimum sampling density corresponds to a sampling matrix of

$$\mathbf{V} = \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix}.$$

The samples' spectrum is a rectangular array of circles packed as shown in Fig. 7.

Example 1. By setting

$$\mathbf{M} = \begin{bmatrix} 1 & 3 \\ 3 & 0 \end{bmatrix},$$

we obtain a decimation geometry that produces a decimation ratio of 1/9. The $\mathbf{k}_{\alpha} = \mathbf{0}$ case is shown in Fig. 8, in which the dashed line outlines a decimation period. From Eq. (6),



Fig. 7. Spectral replications corresponding to the rectangular sampling at the minimum density of a circularly band-limited function.



Fig. 8. Decimation geometry for example 1. A decimation period is outlined by a dashed-line square.

$$\mathbf{U}_d = \begin{bmatrix} 0 & 1/3 \\ 1/3 & -1/9 \end{bmatrix}$$

As shown in Fig. 9, this periodicity matrix produces a subcell in the shape of a distorted hexagon. This distorted hexagonal subcell is totally subsumed in the gaps among the spectral replications when it is centered at (1/2, 1/2). The interpolation kernel to restore the decimated samples is derived, by using Eq. (30), as

$$f_d(t_1, t_2) = \frac{1}{2\sqrt{3}} \operatorname{hinc}\left(\frac{t_1}{3\sqrt{3}} + \frac{t_2}{18\sqrt{3}}, \frac{t_2}{6}\right) \exp[j2\pi(t_1 + t_2)],$$
(31)



Fig. 9. Example 1: a subcell is totally embedded within a gap region.



Fig. 10. Hexagonal support of the Fourier transform of a hinc function.

where hinc (t_1, t_2) is the inverse transform of the unit-amplitude hexagon shown in Fig. 10:

$$\begin{aligned} \operatorname{hinc}(t_1, t_2) &= \frac{4}{\sqrt{3}} \left\{ \operatorname{sinc} \frac{2}{\sqrt{3}} t_1 \operatorname{sinc} 2t_2 + \frac{1}{2\pi t_2} \right. \\ &\times \left[\operatorname{sinc} \frac{1}{\sqrt{3}} (t_1 - \sqrt{3} t_2) \operatorname{sin} \pi (\sqrt{3} t_1 + t_2) \right. \\ &- \operatorname{sinc} \frac{1}{\sqrt{3}} t_1 \operatorname{sin} \pi \sqrt{3} t_1 \right] \end{aligned}$$

where

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$$\operatorname{sinc}(x) = \frac{\sin \pi x}{\pi x}.$$

The phase term in Eq. (31) shifts the hexagonal hinc function into the gap region, as is shown in Fig. 9. We found the work of $Rajan^{12}$ useful in this computation.

Example 2. A 1:8 deletion ratio for the same set of samples can be obtained by choosing

$$\mathbf{M} = \begin{bmatrix} 2 & -2 \\ 2 & 2 \end{bmatrix}$$

The $k_{\alpha} = 0$ case is shown in Fig. 11. The subcell corresponding to the periodicity matrix is



Fig. 11. Decimation geometry for example 2. A decimation period is outlined by a dashed-line rectangle.



Fig. 12. Example 2: a subcell is totally embedded within a gap region.

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as is shown in Fig. 12. The diamond-shaped subcell is shown to be subsumed in the gap when it is centered at (1/2, 1/2), and we compute the interpolation function in the form

$$f_d(t_1, t_2) = \operatorname{sinc}\left(\frac{t_1 + t_2}{4}\right) \operatorname{sinc}\left(\frac{-t_1 + t_2}{4}\right) \exp[j2\pi(t_1 + t_2)].$$
(32)

Effects of Truncating. To illustrate the effects of truncation for both examples 1 and 2, we use the signal $x(t_1, t_2) = \text{jinc}(t_1, t_2)$:

$$\operatorname{jinc}(t_1, t_2) = \frac{2J_1[2\pi(t_1^2 + t_2^2)^{1/2}]}{\pi(t_1^2 + t_2^2)},$$
(33)

where $J_1(x)$ is the Bessel function of the first kind of order $1.^{13}$ The jinc function is also known as the sombero function.¹⁴ The spectrum of this signal is the circle $4\pi \operatorname{rect}[(u_1^2 + u_2^2)^{1/2}/2]$. The value of the sample at the origin, x(0) = 1, was estimated by using windows of various extents. Specifically, the estimations of the decimated samples are

$$\tilde{x}_{d}(\mathbf{V}_{d}\mathbf{m}) = \sum_{i \neq d} \sum_{\mathbf{n} \in \mathcal{M}} x_{i}(\mathbf{V}_{d}\mathbf{n}) f_{d}[\mathbf{V}_{d}(\mathbf{m} - \mathbf{n}) - \mathbf{V}(\mathbf{k}_{i} - \mathbf{k}_{d})],$$
(34)

where

$$\sum_{\mathbf{n}\in\mathcal{M}} = \sum_{n_1=-M}^M \sum_{n_2=-M}^M \dots \sum_{n_N=-M}^M$$

and M is the number of layers used in the estimation. The window extents for the continuation of example 1 of a total of M = 3 layers are shown in Fig. 13, where M = 1 corresponds to 4 decimation periods, M = 2 corresponds to 16 decimation periods, and M = 3 corresponds to 36 decimation periods. The estimate, $\tilde{x}(0)$ is shown versus M in Fig. 14. When only known data are used in the first layer, the sample at the origin is estimated at 0.96. Using M = 2 layers improves the estimate to 1.01, a result within 1% of that desired. For example 2, the estimate $\tilde{x}(0)$ is shown versus M in Fig. 15. Good steady-state convergence occurs at M > 4 layers.

Sampling Below the Nyquist Density: First-Order Decimation of Hexagonal Sampling

We have seen that a circularly band-limited function is sampled at the Nyquist density by using a hexagonal sampling geometry. For a unit-diameter circle, the sampling matrix corresponding to this hexagonal sampling geometry is

$$\mathbf{V} = \begin{bmatrix} 1 & 0\\ -1/\sqrt{3} & 2/\sqrt{3} \end{bmatrix}$$
(35)

The sampling density is D = 0.866. The repetition pattern of the sample's spectrum is shown in Fig. 1.

If the samples can be decimated from this sample set, the function will be sampled below the Nyquist density.

Example 3. Given the hexagonal sampling geometry, we can achieve a 1:64 deletion ratio by adopting a decimation matrix of $V_d = 8V$ (Fig. 16). Thus we obtain

$$\mathbf{U}_{d} = \frac{1}{8} \mathbf{U} = \begin{bmatrix} 1/8 & 1/16 \\ 0 & \sqrt{3}/16 \end{bmatrix}$$

Within a parallelogram cell, we can fit 64 similarly shaped subcells, one of which is subsumed in a gap (Fig. 17). The corresponding interpolation kernel is

$$f_d(t_1, t_2) = \operatorname{sinc} \frac{t_1}{8} \operatorname{sinc} \left(\frac{-t_1 + \sqrt{3}t_2}{16} \right) \\ \times \exp \left[j\pi \left(\frac{-3 + 16\sqrt{3}}{16} t_1 + \frac{\sqrt{3}}{16} t_2 \right) \right]$$
(36)



Fig. 13. Three-layer window corresponding to example 1.



Fig. 14. Plot of the estimate $\tilde{x}(0)$ versus the number of layers for example 1.



Fig. 15. Plot of the estimate $\tilde{x}(0)$ versus the number of layers for example 2.



Fig. 16. Decimation geometry specified in example 3.

The overall sampling density in this example is reduced below the Nyquist density.

Starting from the rectangular sampling geometry, examples 1 and 2 reduce the density to 0.8889 and 0.8750, respectively. The hexagonal sampling geometry yields the Nyquist density of 0.866, and example 3 reduces the density to 0.8525, which is 0.135 below that of Nyquist.

Second-Order Decimation and Restoration

We now extend the result of first-order decimation to second order. If a subcell is small enough that two identically oriented nonoverlapping subcells are totally enclosed within a gap of a single spectral period, then two sample subgroups can be deleted and restored by the remaining L-2 sample subgroups. The result is that the sampling density is reduced by a factor of 2/L. We present this extension by showing that, in example 3, one more subcell can be accommodated within another gap. Therefore an additional sample subgroup can be deleted in example 3 with no loss of information.

Example 4. Continuing with example 3, we can easily position a second subcell located at (1, 1/2), as shown in Fig. 18. We denote this cell $\hat{\mathcal{C}}_{d,2}$ and the first one $\hat{\mathcal{C}}_{d,1}$. Both cells are totally subsumed in the gap. Since $\hat{X}(\nu)$ is identi-



Fig. 17. Example 3: a subcell is totally embedded in a gap region.



Fig. 18. Example 4: two subcells are totally embedded in two gap regions.

cally zero in these two cells, we can obtain two homogeneous equations from Eq. (21):

$$\sum_{i=0}^{L-1} \hat{X}_{i}(\boldsymbol{\nu}) \exp(-2\pi \mathbf{V} \mathbf{k}_{i}) = 0, \quad \boldsymbol{\nu} \in \hat{\mathcal{C}}_{d,1},$$
$$\sum_{i=0}^{L-1} \hat{X}_{i}(\boldsymbol{\nu}) \exp(-2\pi \mathbf{V} \mathbf{k}_{i}) = 0, \quad \boldsymbol{\nu} \in \hat{\mathcal{C}}_{d,2}.$$
(37)

Let the two nonidentical sample subgroups $\{x_k(V_dn)\}|k = k_1, k_2\}$ be deleted. The set of the locations of the deleted samples is therefore

$$\mathcal{M} = \mathcal{M}_1 \cup \mathcal{M}_2, \tag{38}$$

where

$$\mathcal{M}_1 = \{\mathbf{m} | \mathbf{m} = \mathbf{Mn} + \mathbf{k}_1\}, \qquad \mathcal{M}_2 = \{\mathbf{m} | \mathbf{m} = \mathbf{Mn} + \mathbf{k}_2\}.$$

In this case, two samples are deleted per decimation period. The interpolation kernels to restore these two sample subgroups can be derived from Eqs. (37).

Both Eqs. (37), however, are not defined over the same domain. Nevertheless, the kernels can be solved with the addition of some extra steps. By recognizing that the two cells $\hat{\mathcal{C}}_{d,1}$ and $\hat{\mathcal{C}}_{d,2}$ are similarly oriented, we can view $\hat{\mathcal{C}}_{d,2}$ as a translation of $\hat{\mathcal{C}}_{d,1}$ by a vector **d**. We then rewrite Eqs. (37) as

$$\sum_{i=1,2} \exp(-2\pi \boldsymbol{\nu}^T \mathbf{V} \mathbf{k}_i) \hat{X}_i(\boldsymbol{\nu}) = G(\boldsymbol{\nu}), \qquad \boldsymbol{\nu} \in \hat{\mathcal{C}}_{d,1},$$
$$\sum_{i=1,2} \exp[-j2\pi (\boldsymbol{\nu} + \mathbf{d})^T \mathbf{V} \mathbf{k}_i] \hat{X}_i(\boldsymbol{\nu} + \mathbf{d}) = G(\boldsymbol{\nu} + \mathbf{d}),$$
$$\boldsymbol{\nu} \in \hat{\mathcal{C}}_{d,1}, \quad (39)$$

where

$$G(\boldsymbol{\nu}) = -\sum_{j\neq 1,2} \hat{X}_j(\boldsymbol{\nu}) \exp(-j2\pi\boldsymbol{\nu}^T \mathbf{V} \mathbf{k}_j).$$
(40)

Every subgroup's spectrum is periodic with the periodicity matrix, \mathbf{U}_d ; i.e., $\hat{x}_j(\mathbf{v}) = \hat{x}_j(\mathbf{v} + \mathbf{U}_d \mathbf{p})$ for any integer vector \mathbf{p} . Hence, if \mathbf{d} is an integer-weighted sum of the two periodicity vectors \mathbf{u}_{d_1} and \mathbf{u}_{d_2} (the two column vectors of \mathbf{U}_d in example 3), then $\hat{X}_j(\mathbf{v} + \mathbf{d}) = \hat{X}_j(\mathbf{v})$. The two equations above can be combined into a matrix equation:

$$\mathbf{H}_{d}\mathbf{X}_{d} = \mathbf{G}, \qquad \mathbf{\nu} \in \hat{\mathcal{C}}_{d,1}, \tag{41}$$

where

$$\begin{split} \mathbf{H}_{d} &= \begin{bmatrix} \exp(-j2\pi\nu^{T}\mathbf{V}\mathbf{k}_{1}) & \exp(-j2\pi\nu^{T}\mathbf{V}\mathbf{k}_{2}) \\ \exp[-j2\pi\nu^{T}(\mathbf{V}+\mathbf{d})\mathbf{k}_{1}] & \exp[-j2\pi\nu^{T}(\mathbf{V}+\mathbf{d})\mathbf{k}_{2}] \end{bmatrix}, \\ \mathbf{X}_{d} &= \begin{bmatrix} \hat{X}_{\mathbf{k}_{1}}(\boldsymbol{\nu}) \\ \hat{X}_{\mathbf{k}_{2}}(\boldsymbol{\nu}) \end{bmatrix}, \\ \mathbf{G} &= \begin{bmatrix} G(\boldsymbol{\nu}) \\ G(\boldsymbol{\nu}+d) \end{bmatrix}, \end{split}$$

and

$$G(\mathbf{\nu} + \mathbf{d}) = -\sum_{j \neq 1,2} \hat{X}_j(\mathbf{\nu}) \exp[-j2\pi(\mathbf{\nu} + \mathbf{d})^T \mathbf{V} \mathbf{k}_j].$$
(42)

The vector G can also be written as

 $\mathbf{G} = -\mathbf{H}_r \mathbf{X}_r$

where X_r is the vector whose entries are the spectra of the remaining sample subgroups and H_r is the matrix whose entries are the weights of the remaining sample subgroups as specified in Eqs. (40) and (42). If the matrix H_d is nonsingular, then the spectra of the two deleted sample subgroups can be obtained:

$$\mathbf{X}_d = -\mathbf{H}_d^{-1}\mathbf{H}_r\mathbf{X}_r \tag{43}$$

or

$$\hat{X}_{i}(\mathbf{v}) = -\sum_{j \neq 1,2} H_{i,j}(\mathbf{v}) \hat{X}_{j}(\mathbf{v}), \quad i = 1, 2,$$

where the $\mathbf{H}_{i,j}(\boldsymbol{\nu})$'s are computed from the matrix product $-\mathbf{H}_d^{-1}\mathbf{H}_r$. In particular, let

$$H_{i,j}(\boldsymbol{\nu}) = C_{i,j}(\mathbf{d}) \exp[-j2\pi\boldsymbol{\nu}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_i)],$$

where

$$C_{1,j}(\mathbf{d}) = -\frac{\sin[\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_2)]}{\sin[\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_1 - \mathbf{k}_2)]} \exp[-j\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_1)],$$

and let

$$H_{2,j} = C_{2,j}(\mathbf{d}) \exp[-j2\pi \boldsymbol{\nu}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_2)],$$

where

$$C_{2,j}(\mathbf{d}) = \frac{\sin[\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_1)]}{\sin[\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_1 - \mathbf{k}_2)]} \exp[-j\pi \mathbf{d}^T \mathbf{V}(\mathbf{k}_j - \mathbf{k}_2)].$$

The two $C_{i,j}(\mathbf{d})$'s are constant for a given \mathbf{d} and \mathbf{k}_j . The samples of the two deleted subgroups can be obtained by the following interpolation formula:

$$x_i(\mathbf{V}_d \mathbf{m}) = -\sum_{j \neq 1,2} \sum_{\mathbf{n}} x_j(\mathbf{V}_d \mathbf{n}) h_{i,j} [\mathbf{V}_d(\mathbf{m} - \mathbf{n}) - \mathbf{V}(\mathbf{k}_j - \mathbf{k}_i)],$$

$$i = 1, 2, \quad (44)$$

where

$$h_{i,j}(\mathbf{t}) = \frac{1}{D_d} C_{i,j}(\mathbf{d}) \int_{\hat{\mathcal{C}}_{d,1}} \exp(j2\pi \boldsymbol{\nu}^T \mathbf{t}) d\boldsymbol{\nu}$$

is the interpolation kernel.

Unfortunately, **d** in this example is not an integer-weighted sum of the two periodicity vectors. We found that

$$\mathbf{d} = \left(\frac{7}{16}, \frac{16 - 7\sqrt{3}}{16}\right)^T \approx 2.38\mathbf{u}_{d_1} + 2.24\mathbf{u}_{d_2}$$

In order to apply the results above, we need to add some additional steps. First, we divide the cells $\hat{\mathcal{C}}_{d,1}$ into four partitions: $\hat{\mathcal{C}}_{d,1,k}$, k = 1, 2, 3, 4. Likewise for the cells $\hat{\mathcal{C}}_{d,2}$, we have $\hat{\mathcal{C}}_{d,2,k}$, k = 1, 2, 3, 4. We partition both cells in a way

such that for every $\mathbf{v}_1 \in \hat{\mathcal{C}}_{d,1,k}$ there exists a corresponding $\mathbf{v}_2 \in \hat{\mathcal{C}}_{d,2,k}$ such that

$$v_2 = v_1 + \mathbf{d}_k, \qquad k = 1, 2, 3, 4.$$

Normally, the number of partitions for an N-dimensional signal is 2^N . Each vector \mathbf{d}_k is an integer-weighted sum of the two periodicity vectors:

$$\mathbf{d}_k = \mathbf{U}_d \mathbf{p}_k, \qquad k = 1, 2, 3, 4,$$

where \mathbf{p}_i is an integer vector for every *i*. The four \mathbf{p}_i vectors are measured to be

$$p_1 = (3, 3)^T$$
, $p_2 = (2, 3)^T$, $p_3 = (3, 2)^T$, $p_4 = (2, 2)^T$.

We illustrate all these measurements in Fig. 19. Since we can view $\hat{\mathcal{C}}_{d,2,k}$ as a translation of $\hat{\mathcal{C}}_{d,1,k}$ by the vector \mathbf{d}_k , and also $X(\mathbf{v}) = X(\mathbf{v} + \mathbf{d}_k)$, we can solve Eq. (41) over the four regions $\hat{\mathcal{C}}_{d,1,k}$ and obtain four solutions. The restoration formula for the two deleted subgroups is then a combination of the union of the four solutions.

Recall the procedure starting at Eq. (43). By substituting each vector \mathbf{d}_k into Eq. (41), we obtain four matrix equations, each corresponding to a partition $\hat{\mathcal{C}}_{d,1,k}$:

$$\mathbf{H}_{d_k} \mathbf{X}_{d_k} = \mathbf{G}_k, \quad \nu \in \hat{\mathcal{O}}_{d,1,k}, \quad k = 1, 2, 3, 4.$$
(45)

Since

$$\mathbf{X}_d = \bigcup_{k=1}^4 \mathbf{X}_{d_k},$$

it follows that, if every matrix \mathbf{H}_{d_k} is invertible, then

$$\mathbf{X}_d = \bigcup_{k=1}^4 \mathbf{H}_{d_k}^{-1} \mathbf{G}_k.$$

The interpolation kernels can then be obtained:



Fig. 19. Example 4: an illustration of the partitioning of each of the two subcells into four partitions.

$$h_{i,j}(\mathbf{t}) = \frac{1}{D_d} \sum_{k=1}^{4} C_{i,j}(\mathbf{d}_k) h_k(\mathbf{t}), \qquad i = 1, 2,$$
(46)

where

$$h_k(\mathbf{t}) = \int_{\hat{\mathcal{C}}_{d,1,k}} \exp(j2\pi\nu^T \mathbf{t}) \mathrm{d}\nu.$$
(47)

Now we evaluate Eq. (47) at each partition $\hat{\mathcal{C}}_{d,1,k}$. Since all the partitions are similar parallelograms but are different in scale and location, Eq. (47) can be evaluated over a parallelogram region, say, a subcell. We then evaluate $h_k(\mathbf{t})$ for every k by applying the corresponding scaling and shifting operations. Specifically, we first evaluate Eq. (47) over a subcell \mathcal{C}_d with its lower left-hand corner located at the origin. The solution of this integration is denoted as a general solution $k(t_1, t_2)$, which is found to be

$$k(t_1, t_2) = \frac{\sqrt{3}}{128} \operatorname{sinc}\left(\frac{t_1}{8}\right) \operatorname{sinc}\left(\frac{t_1}{16} + \frac{\sqrt{3}t_2}{16}\right) \\ \times \exp\left[j\pi\left(\frac{3t_1}{16} + \frac{\sqrt{3}t_2}{16}\right)\right]. \quad (48)$$

We consider the scaling operation first. The scaling between each partition $\hat{\mathcal{C}}_{d,1,k}$ and \mathcal{C}_{d0} is represented by a scaling matrix, denoted \mathbf{S}_k . Since each partition is in the same orientation as the subcell, \mathbf{S}_k is a diagonal matrix:

$$\mathbf{S}_{k} = \begin{bmatrix} s_{k1} & 0\\ 0 & s_{k2} \end{bmatrix}, \qquad \dot{k} = 1, 2, 3, 4,$$

where s_{k1} and s_{k2} are the scaling factors in the direction of \mathbf{u}_{d_1} and \mathbf{u}_{d_2} , respectively. By incorporating the shifting vector, we obtain the relationship

$$\hat{\mathcal{C}}_{d,1,k}(\boldsymbol{\nu}) = \mathcal{C}_{d0}(\mathbf{S}_k \boldsymbol{\nu} + \mathbf{r}_k), \qquad k = 1, 2, 3, 4,$$

where \mathbf{r}_k is the displacement vectors of $\hat{\mathcal{C}}_{d,1,k}$ in relation to \mathcal{C}_{d0} . By applying Rajan's result,¹² we obtain

$$h_k(\mathbf{t}) = s_{k1} s_{k2} k(\mathbf{S}_k^T \mathbf{t}) \exp(j2\pi \mathbf{r}_k^T \mathbf{t}), \qquad (49)$$

where k(t) is the function expressed in Eq. (48). The measurements of S_k and r_k for each partition are listed in Table 1. Finally, we substitute the corresponding values listed in the table into Eq. (49); the result is then substituted into Eq. (46). We obtain the following interpolation kernels for restoring the two deleted sample subgroups:

$$h_{i,j}(\mathbf{t}) = \frac{1}{D_d} \sum_{k=1}^{4} C_{i,j} s_{k1} s_{k2} k(\mathbf{S}_k^T \mathbf{t}) \exp(j2\pi \mathbf{r}_k^T \mathbf{t}).$$
(50)

In this example we obtain a sampling density of 0.8389, the lowest so far in all the examples.

A sufficient condition for the existence of the solution to restore the deleted samples is that all matrices, \mathbf{H}_{d_k} in Eq. (45), are nonsingular. One can show, for example, that a solution exists for the case of $\mathbf{k}_1 = (0, 0)^T$ and $\mathbf{k}_2 = (1, 1)^T$. If \mathbf{k}_2 is changed to $(1, 0)^T$, the condition is violated, and therefore no solution exists for this decimation geometry. In contrast to the first-order decimation, second-order decimation does not guarantee that solutions exist for all possible decimation geometries.

k	<i>s</i> _{k1}	s_{k2}	r_{k1}	r_{k2}
1	$\frac{10\sqrt{3}-16}{2\sqrt{3}}$	$\frac{16-9\sqrt{3}}{\sqrt{3}}$	$\frac{7}{16}$	$\frac{7\sqrt{3}-8}{10}$
2	$\frac{8-4\sqrt{3}}{\sqrt{3}}$	$\frac{16-9\sqrt{3}}{\sqrt{3}}$	$\frac{17\sqrt{3}-16}{16\sqrt{3}}$	$\frac{7\sqrt{3}-8}{10}$
3	$\frac{5\sqrt{3}-8}{\sqrt{3}}$	$\frac{10\sqrt{3}-16}{\sqrt{3}}$	$\frac{8-\sqrt{3}}{8\sqrt{3}}$	$\frac{4\sqrt{3}-3}{8\sqrt{3}}$
4	$\frac{8-4\sqrt{3}}{\sqrt{3}}$	$\frac{10\sqrt{3}-16}{\sqrt{3}}$	$\frac{1}{2}$	$\frac{4\sqrt{3}-3}{8\sqrt{3}}$

Multiple-Order Decimation and Restoration

The interpolation kernels in Eq. (46) have a general form of

$$f_{i,j}(\mathbf{t}) = \sum_{k=1}^{4} a_{i,j,k} f_k(\mathbf{t}).$$
 (51)

By comparing this with Eq. (46), we obtain

$$a_{i,j,k} = C_{i,j}(\mathbf{d}_k)$$

and

$$f_k(\mathbf{t}) = \frac{1}{D_d} s_{k1} s_{k2} k(\mathbf{S}_k^T \mathbf{t}) \exp(j2\pi \mathbf{r}_k^T \mathbf{t}).$$

In this form, the kernels are determined by the functions $C_{i,j}(\mathbf{d})$, which are obtained by inverting the matrix \mathbf{H}_d in Eq. (41). The values of the \mathbf{s}_k 's and \mathbf{r}_k 's can be determined immediately after these factors are known.

With this approach, we consider extending the decimation to higher orders. By the same reasoning, if a subcell is small enough that q > 2 identically oriented subcells without overlap are totally enclosed within gaps of a single spectral period, the q of L sample subgroups can be deleted and restored by the L - q remaining sample subgroups. Specifically, let $\{\mathbf{u}_{d_i} | i = 1, \ldots, N\}$ be the N periodicity vectors, and let $\hat{\mathcal{C}}_{d,j}, j$ $= 1, \ldots, q$, be the q subcells enclosed within the gap regions. If we choose $\hat{\mathcal{C}}_{d,1}$ to be the reference subcell, the locations of the other q - 1 subcells are identified by the following displacement notations:

$$\hat{\mathcal{C}}_{d,i} = \{ \boldsymbol{\nu} | \boldsymbol{\nu} \in \{ \hat{\mathcal{C}}_{d,1} \oplus \mathbf{d}_j \} \}, \qquad j = 2, \ldots, q,$$

where \oplus denotes offset by. After the location of every subcell is identified, the next step is consideration of the partitioning of the subcells. In general, each subcell introduces an extra partitioning boundary in every \mathbf{u}_{d_i} direction. For qsubcells in a N-dimensional space, then, the total number of partitions is q^N . After the partitioning, we have, for every partition $\hat{\mathcal{C}}_{d,1,k}$ in $\hat{\mathcal{C}}_{d,1}$, a corresponding partition $\hat{\mathcal{C}}_{d,j,k}$ in every *j*th subcell such that

$$\hat{\mathcal{C}}_{d,j,k} = \{ \boldsymbol{\nu} | \boldsymbol{\nu} \in \{ \hat{\mathcal{C}}_{d,1,k} \oplus \mathbf{d}_{j,k} \} \},\$$

$$j = 2, \dots, q, \qquad k = 1, \dots, q^N,$$

where $\mathbf{d}_{j,k}$ is the displacement vector of the kth partition in the *j*th subcell relative to the kth partition in the reference subcell $\hat{\mathcal{C}}_{d,1,k}$,

$$\mathbf{d}_{i,k} = \mathbf{U}_d \mathbf{p}_{i,k},$$

and $\mathbf{p}_{j,k}$ is an integer vector. Also, the dimension of the $\hat{\mathcal{C}}_{d,1,k}$ is obtained through the scaling of the reference subcell, $\hat{\mathcal{C}}_{d,1}$. The scaling factor in every \mathbf{u}_{d_i} direction of the *k*th partition is denoted $s_{k,i}$, $i = 1, \ldots, N$. The displacement vectors, $\mathbf{d}_{j,k}$, and the scaling factors, \mathbf{s}_k , can be obtained geometrically.

Let \mathcal{M} be the set of the indices of the q decimated sample subgroups. In a form similar to Eq. (41), we obtain, in every partition,

$$\mathbf{H}_{d,k}\mathbf{X}_{d} = \mathbf{G}, \qquad \mathbf{v} \in \hat{\mathcal{C}}_{d,1,k},$$

where the matrix $\mathbf{H}_{d,k}$ has the entries $\{\exp[-j2\pi(\nu + \mathbf{d}_{j,k})^T \mathbf{V} \mathbf{k}_i] | i \in \hat{\mathcal{M}}\}$. Also, we let

$$k(\mathbf{t}) = \int_{\hat{\mathcal{C}}_{d,1}} \exp(j2\pi\nu^T \mathbf{t}) \mathrm{d}\nu$$

The interpolation kernels follow:

$$f_{i,j}(\mathbf{t}) = \sum_{k=1}^{q^N} a_{i,j,k} f_k(t), \qquad i \in \hat{\mathcal{M}}, \qquad j \in \hat{\mathcal{M}}, \tag{52}$$

where $a_{i,j,k}$ is obtained from the inversion of the matrix $\mathbf{H}_{d,k}$, and

$$f_k(\mathbf{t}) = \frac{1}{D_d} \left(\prod_{n=1}^N S_{k,n} \right) k(\mathbf{S}_k^T \mathbf{t}) \exp(j2\pi \mathbf{r}_k^T \mathbf{t}).$$

The vector \mathbf{r}_k is the displacement vector of $\hat{\mathcal{C}}_{d,1,k}$ relative to $\hat{\mathcal{C}}_{d,1}$. The cardinal series is then

$$x_{i}(\mathbf{V}_{d}\mathbf{m}) = -\sum_{j \notin \hat{\mathcal{M}}} \sum_{\mathbf{n}} x_{j}(\mathbf{V}_{d}\mathbf{n})h_{i,j}[\mathbf{V}_{d}(\mathbf{m}-\mathbf{n}) - \mathbf{V}(\mathbf{k}_{j}-\mathbf{k}_{i})],$$
$$i \in \hat{\mathcal{M}}.$$
(53)

Again, a sufficient condition for the existence of a solution is that all q^N matrices, $\mathbf{H}_{d,k}$, be nonsingular. As is evident from the derivation, complexity increases with the order of decimation and the dimension of the function.

EFFECTS OF DATA NOISE

We now consider the effects of data noise on the recovery of the decimated samples. The data noise is modeled as discrete white noise.¹ The discrete-white-noise model was used previously to represent the effect of quantization error¹⁵ and the effect of sampling position jitter.¹⁶ We can show that the interpolation noise level (INL) is the data noise level multiplied by an amplification ratio, which is referred to as the noise-level amplification ratio (NLAR).

Interpolation Noise Level of First-Order Decimation

Let $\{\xi(\mathbf{Vn}) | \mathbf{n} \notin \mathcal{M}\}$, the data noise, be a zero-mean stationary discrete white process,

$$E\{\xi(\mathbf{Vn})\xi(\mathbf{Vm})\} = \xi^2 \delta[\mathbf{n} - \mathbf{m}],$$

where E denotes expectation, $\overline{\xi^2} = E[\xi^2(t)]$ is the data noise level, and $\delta[\cdot]$ is the Kronecker delta (note the square brackets). Let the interpolation noise be $\eta(t)$. With reference to Eq. (29), the interpolation noise at the decimation lattice points is

$$\eta(\mathbf{V}_{d}\mathbf{n}) = \sum_{i \neq d} \sum_{\mathbf{m}} \xi(\mathbf{V}_{d}\mathbf{m} + \mathbf{V}\mathbf{k}_{i})\mathbf{f}_{d}[\mathbf{V}_{d}(\mathbf{n} - \mathbf{m}) - (\mathbf{k}_{i} - \mathbf{k}_{a})],$$
$$\mathbf{n} \in \mathcal{M}.$$
(54)

Again, we let the offset vector $\mathbf{k}_{\alpha} = \mathbf{0}$. Since the data noise is stationary and the deletion is periodic, the restoration noise statistics will be the same at every point. Thus it suffices to examine the restoration noise at the origin only. Since

$$\eta(\mathbf{0}) = \sum_{i \neq 0} \sum_{\mathbf{m}} \xi(\mathbf{V}_d \mathbf{m} + \mathbf{V} \mathbf{k}_i) f_d(-\mathbf{V}_d \mathbf{m} - \mathbf{V} k_i)$$
(55)

and

 $E\{\xi(\mathbf{V}_d\mathbf{m} + \mathbf{V}\mathbf{k}_r)\xi^*(\mathbf{V}_d\mathbf{n} + \mathbf{V}\mathbf{k}_s)\}$ = $\overline{\xi^2}\delta[\mathbf{n} - [\mathbf{n} + \mathbf{M}^{-1}(\mathbf{k}_r - \mathbf{k}_s)]],$

then squaring both sides of Eq. (55) and expectation gives the expression for the INL:

$$\overline{\eta^{2}} = \overline{\xi^{2}} \sum_{r \neq 0} \sum_{s \neq 0} \sum_{\mathbf{m}} \sum_{\mathbf{n}} \delta[\mathbf{n} - [\mathbf{m} + \mathbf{M}^{-1}(\mathbf{k}_{r} - \mathbf{k}_{s})]]$$

$$\times f_{d}(\mathbf{V}_{d}\mathbf{m} - \mathbf{V}\mathbf{k}_{r})f_{d}^{*}(\mathbf{V}_{d} - \mathbf{V}\mathbf{k}_{s})$$

$$= \overline{\xi^{2}} \sum_{r \neq 0} \left[\sum_{\mathbf{m}} |f_{d}(\mathbf{V}_{d}\mathbf{m} - \mathbf{V}\mathbf{k}_{r})|^{2} \right].$$
(56)

The sum inside the large brackets can be evaluated by Parseval's theorem. For every r from 0 to L - 1,

$$\sum_{\mathbf{m}} |f_d (\mathbf{V}_d \mathbf{m} - \mathbf{V} \mathbf{k}_r)|^2 = D_d \int_{\hat{\mathcal{C}}_d} |F_d(\boldsymbol{\nu})|^2 \mathrm{d}\boldsymbol{\nu}, \tag{57}$$

where $F_d(\mathbf{v})$ is the Fourier spectrum of the kernel $f_d(\mathbf{t})$. With reference to Eq. (30),

$$|F_d(\mathbf{v})| = \begin{cases} 1/D_d & \mathbf{v} \in \hat{\mathcal{C}}_d \\ 0 & \text{otherwise} \end{cases}$$

By substituting the value into Eq. (57) and also because the area of $\hat{\mathcal{C}}_d$ is equal to D_d , we obtain

$$\sum_{\mathbf{m}} |f_d (\mathbf{V}_d \mathbf{m} - \mathbf{V} \mathbf{k}_r)|^2 = 1,$$

and the interpolation noise level in Eq. (56) is obtained:

$$\overline{\eta^2} = (L-1)\overline{\xi^2}.$$
(58)

The ratio $\overline{\eta^2/\xi^2}$, defined as the NLAR, is equal to L - 1. Clearly, this ratio is the same regardless of the sampling geometry and the decimation geometry. The NLAR's for examples 1-3 of the first-order sample decimation are therefore 8, 7, and 63, respectively.

Interpolation Noise Level of Multiple-Order Decimation For multiple-order decimation, the INL's for different decimation subgroups are different. Briefly stated, the NLAR for the *i*th subgroup is

$$\frac{\overline{\eta_i^2}}{\overline{\xi^2}} = \sum_{j \notin \mathcal{M}} \sum_{k=1}^{q^N} |a_{i,j,k}|^2 \left(\prod_{n=1}^N s_{kn}\right), \quad i \in \hat{\mathcal{M}}.$$
 (59)

For first-order decimation, the coefficients $\{a_{i,j,k}\}\$ and $\{s_{kn}\}\$ are equal to 1, and therefore Eq. (59) is reduced to Eq. (58). For the multiple-order case, the first set of coefficients is generally different for different *i*'s and thus for the NLAR's. For our example 4, the NLAR's for the two subgroups, $\mathbf{k}_1 = (0, 0)^T$ and $\mathbf{k}_2 = (1, 1)^T$, are identical and equal to 36.3012.

Normally, the magnitudes of the $a_{i,j,k}$'s are higher, and hence the NLAR's are higher, in cases in which the deleted subgroups are more densely clustered.^{7,8} Such an effect of the NLAR versus the varying degrees of clustering was demonstrated by Yen⁸ and Marks and Radbel.^{17–19} We therefore need an algorithm to search for the combination of decimated subgroups for which the NLAR's are the lowest and most uniform of all possible combinations. The NLAR will generally be lower for higher-order decimation because the number of terms in Eq. (59) decreases with the order of decimation. Such an effect can be seen from the different NLAR's in examples 3 and 4.

CONCLUSION

In this paper we have demonstrated a technique to sample directly certain multidimensional bandlimited functions below the Nyquist density. The technique is implemented by sample decimation and can be applied directly to bandlimited functions of arbitrary dimension and arbitrary support using arbitrary nonaliasing sampling geometries. The decimated samples are restored by discrete interpolation. The interpolation kernels in many cases can be obtained in closed form. The INL caused by interpolation from noisy data was also analyzed. The magnitude of the INL decreases with the order of decimation and the uniformity of the decimation lattice.

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