Superposition Frames for Adaptive Time-Frequency Analysis and Fast Reconstruction

Daniel Rudoy, Student Member, IEEE, Prabahan Basu, Member, IEEE, and Patrick J. Wolfe, Senior Member, IEEE

Abstract—In this article we introduce a broad family of adaptive, linear time-frequency representations termed superposition frames, and show that they admit desirable fast overlap-add reconstruction properties akin to standard short-time Fourier techniques. This approach stands in contrast to many adaptive time-frequency representations in the existing literature, which, while more flexible than standard fixed-resolution approaches, typically fail to provide for efficient reconstruction and often lack the regular structure necessary for precise frame-theoretic analysis. Our main technical contributions come through the development of properties which ensure that our superposition construction provides for a numerically stable, invertible signal representation. Our primary algorithmic contributions come via the introduction and discussion of specific signal adaptation criteria in deterministic and stochastic settings, based respectively on time-frequency concentration and nonstationarity detection. We conclude with a short speech enhancement example that serves to highlight potential applications of our approach.

Index Terms—Adaptive short-time Fourier analysis, frame theory, Gabor frames, overlap-add synthesis, speech enhancement.

I. INTRODUCTION

OVERCOMPLETE short-time Fourier methods are frequently used to analyze the time-varying spectral content of discrete-time waveforms \( x[t] \) arising in a variety of signal processing applications. Since the choice of localizing window function effectively controls the balance between time and frequency resolution \( a \ priori \), standard representations cannot modulate this trade-off to adapt to the local spectral content of \( x[t] \). Over the past two decades, this shortcoming has motivated the development of various linear and nonlinear adaptive time-frequency analysis methods [2]–[14], in applications ranging from biomedical engineering [15] to radar signal analysis [16] and speech processing [17].

Despite the recognized importance of overcomplete signal-adaptive time-frequency analysis, the above methods generally fail to admit fast reconstruction of the signal \( x[t] \) from its time-frequency representation, by which we mean any non-iterative method that avoids direct (pseudo-)inversion of the corresponding analysis operator. While approaches such as modulated lapped transforms for audio coding [18], wavelet packet decompositions via best basis [19], [20] and adaptive segmentation via dynamic programming [21]–[25] can lead to flexible tilings of the time-frequency plane, the general goal of efficient reconstruction from signal-adaptive, overcomplete time-frequency representations remains an open problem. This issue is particularly important, given the recent interest in oversampled, modulated filter banks [26]–[28].

In this article we introduce a broad family of adaptive, linear time-frequency representations that admit a fast overlap-add reconstruction property akin to standard short-time Fourier techniques. We do so by adapting a given discrete Gabor frame to an observed signal \( x[t] \) via superpositions of neighboring translates of a single window function, to yield the superposition frames of the article title. Related procedures include the multi-window constructions of [29], in which multiple systems are defined on the same time-frequency lattice; and the multi-Gabor expansions of [30], in which multiple time lattices and windows are employed. However, neither of these schemes treats the use of subset selection to achieve a signal-adaptive system in the manner of the present article. More recent approaches [9]–[12] address subset selection from a Gabor frame or union of Gabor frames, but do not consider the structure of the corresponding canonical dual.

A very recent approach in this direction is the study of general nonstationary Gabor frames [31], and indeed our contribution can be viewed as one possible instantiation of this framework. However, as we show below, the additional structure induced by our superposition construction yields several important properties, including, among other results, a preservation of the lower frame bound of the original Gabor frame, a generalized constant overlap-add property that avoids the explicit computation of dual windows, and a means of generating new families of adaptive lapped frames.

The article is organized as follows. We begin by reviewing the short-time Fourier transform and Gabor systems on \( \mathbb{C}^L \) in Section II. Next, we introduce superposition windows in Section III and use them to construct superposition systems in Section IV. In Section V we prove that the resultant systems are in fact frames for \( \mathbb{C}^L \) and study their frame-theoretic properties, and in Section VI we establish fast reconstruction via an analysis of the corresponding frame operator. In Section VII we give examples of signal-dependent adaptation algorithms and illustrate their application to superposition systems. We conclude with a brief discussion in Section VIII.

II. PRELIMINARIES

We first review some well-known properties of Gabor frames [28], [32], [33] and discuss their relationship to short-time Fourier analysis. We take as our setting the space \( \mathbb{C}^L \), and interpret its members as discrete-time \( L \)-periodic signals \( x \in \ell^2(\mathbb{Z}_L) \), with \( \mathbb{Z}_L \) denoting the integers \( \mathbb{Z} \) modulo \( L \). The short-time Fourier transform (STFT) on \( \mathbb{C}^L \) uses a well-concentrated window function in order to localize \( x \) in time prior to the analysis of its frequency content.
Definition 1 (Short-Time Fourier Transform): Fix a window \( w \in \mathbb{C}^L \) and time-frequency lattice constants \( a, b > 0 \) that divide \( L \), with \( a \) an integer, and define \( M, N : N a = M b = L \). Then for the \( n \)th frequency bin index and \( n \)th window shift, with \( m \in \mathbb{Z}_M \) and \( n \in \mathbb{Z}_N \), the Gabor or subsampled short-time Fourier transform \( X[m, n] \) of \( x \in \mathbb{C}^L \) is given by

\[
X[m, n] \triangleq \sum_{t=0}^{L-1} x[t] w[t - na] e^{2 \pi i mbt/L},
\]

where \( i = \sqrt{-1} \) and \( \bar{\tau} \) denotes complex conjugation. The expression of (1) can be viewed as a set of inner products of \( x \) with \( N M \) time-frequency shifts of the chosen window \( w \). To realize this correspondence, and to set notation, we introduce explicit translation and modulation operators as follows.

Definition 2 (Translation and Modulation Operators): Let the translation and modulation operators \( T \) and \( M \) be defined as maps from \( \mathbb{C}^L \) to itself acting according to:

\[
T_{na} w[t] \triangleq w[t - na], \quad M_{mb} w[t] \triangleq w[t] e^{2 \pi i mbt/L}.
\]

Through the action of these operators, time-frequency shifts of the chosen window \( w \in \mathbb{C}^L \) may be indexed as

\[
\phi_{m,n}[t] \triangleq M_{mb} T_{na} w[t], \quad m \in \mathbb{Z}_M, n \in \mathbb{Z}_N,
\]

and one speaks of a Gabor system \( \mathcal{G}(w, a, b) = \{\phi_{m,n}\} \). In order to ensure a reconstruction property for any \( x \) from its subsampled short-time Fourier transform \( X[m, n] \), the Gabor system \( \mathcal{G}(w, a, b) \) must form a frame for \( \mathbb{C}^L \) as follows.

Definition 3 (Gabor Systems and Frames): A denumerable set \( \{\phi_{m,n}\} \) of vectors comprising time-frequency shifts of a single window function \( w \in \mathbb{C}^L \) is called a Gabor system, and is said to be a Gabor frame for \( \mathbb{C}^L \) if there exist constants \( 0 < A \leq B < \infty \) such that

\[
\forall x \in \mathbb{C}^L, A ||x||^2 \leq \sum_{m,n} |\langle x, \phi_{m,n} \rangle|^2 \leq B ||x||^2,
\]

with inner product \( \langle x, \phi_{m,n} \rangle \triangleq \sum_{t=0}^{L-1} x[t] \bar{\phi}_{m,n}[t] = X[m,n] \).

An upper frame bound \( B \) for (3) is guaranteed whenever the set \( \{\phi_{m,n}\} \) is finite, and so the existence of a lower frame bound \( A > 0 \), for a finite Gabor system \( \mathcal{G}(w, a, b) \), is equivalent to the requirement that its elements span \( \mathbb{C}^L \). This occurs if and only if the frame operator is of full rank.

Definition 4 (Gabor Frame Operator): Let \( \mathcal{G}(w, a, b) = \{\phi_{m,n}\} \) be a Gabor system on \( \mathbb{C}^L \), and define the frame operator \( S : \mathbb{C}^L \to \mathbb{C}^L \) through its action on \( x \) as \( Sx = \sum_{m,n} \langle x, \phi_{m,n} \rangle \phi_{m,n} \). Then \( S \) is represented by the \( L \times L \) symmetric and positive semi-definite matrix with entries

\[
S[t, t'] \triangleq \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} M_{mb} T_{na} w[t] \bar{M}_{mb} T_{na} w[t'],
\]

Remark 1 (Strict Positive-Definiteness of Frame Operator): By Definition 4 the frame condition of (3) is equivalent to strict positive definiteness of \( S \) and hence a necessary condition is that \( MN \geq L \) (i.e., \( ab \leq L \)). Moreover, the minimal and maximal eigenvalues of \( S \) yield optimal frame bounds, since (3) may be expressed as the requirement that \( A(x,x) \leq \langle Sx, x \rangle \leq B(x,x), \forall x \in \mathbb{C}^L \).

The frame condition of (3) in turn implies the following reconstruction property:

\[
\forall x \in \mathbb{C}^L, t \in \mathbb{Z}_L, x[t] = \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} \langle x, \phi_{m,n} \rangle \bar{\phi}_{m,n}[t],
\]

where the elements \( \{\phi_{m,n}\} \) comprise a (not necessarily unique) dual frame. However, to each frame may be associated a unique canonical dual, whose elements are given by the action of the frame operator inverse \( S^{-1} \) on each \( \phi_{m,n} \). Moreover, in the Gabor setting, this canonical dual takes the form of another Gabor system \( \mathcal{G}(\bar{w}, a, b) \), with \( \bar{w} \triangleq S^{-1} w \).

Any \( S \) can be written as a sum of outer products of each frame vector with itself; from (4) via the orthogonality relation

\[
\sum_{m=0}^{M-1} \sum_{n=0}^{N-1} e^{2 \pi i mb(t-t')} = \begin{cases} M & \text{when } M \text{ divides } t-t', \\ 0 & \text{otherwise}, \end{cases}
\]

we obtain the so-called Walnut representation [34] of a Gabor frame operator \( S \), which will be used repeatedly throughout.

Definition 5 (Discrete Walnut Representation): Denote by \( M \{t-t'\} \) the condition that \( M \) divides \( t-t' \), and by \( \mathbb{I}_M \{t-t'\} \) the corresponding indicator function on \( \mathbb{Z}_L \). Then the frame operator \( S \) of a finite Gabor system \( \mathcal{G}(w, a, b) \) has banded structure, and satisfies the entrywise relation

\[
S[t, t'] = \mathbb{I}_M \{t-t'\} \cdot M \sum_{n=0}^{N-1} T_{na} w[t] T_{na} w[t'],
\]

Remark 2 (Covering Condition): Note that if \( \mathcal{G}(w, a, b) \) is a frame for \( \mathbb{C}^L \), then (5) implies that the covering condition

\[
\sum_{n=0}^{N-1} ||w[t-na]||^2 > 0, \forall t \in \mathbb{Z}_L
\]

must be satisfied, since a necessary condition for positive definiteness of \( S \) is that its diagonal entries are positive.

Remark 3 (Window Length as Distinct from Support): The support of \( w \in \mathbb{C}^L \) refers to the set of indices \( t \) for which \( w[t] \neq 0 \), with \( ||\text{supp}(w)|| \) its cardinality. Bearing in mind the summands of (5) and (6), define the length of \( w \) by

\[
\text{len}(w) \triangleq ||\text{supp}(w)||
\]

if \( \text{supp}(w) \) is contiguous, as is often the case in practice, and \( \min_{n \in \mathbb{Z}_L} \{\max_{t, t' \in \mathbb{Z}_L} |t-t'+1| : T_{na} w[t] T_{na} w[t'] \neq 0 \} \) otherwise.

Remark 4 (Diagonal Frame Operator): It follows from (5) and (7) that \( S \) is diagonal if \( M \geq \text{len}(w) \), since \( T_{na} w[t] T_{na} w[t'] = 0 \) for all \( t-t' \geq \text{len}(w) \), including those for which \( M \) divides \( t-t' \). In turn, this implies efficient computation of the dual frame \( \{M_{mb} T_{na} \bar{w}\} \), with \( \bar{w} = S^{-1} w \) obtained via element-wise division of \( w \) by \( S[t, t] = M \sum_{n=0}^{N-1} ||w[t-na]||^2 \). In this case the condition of (6) is sufficient to guarantee the frame condition of (3).

We conclude by using the arguments of Remarks 3 and 4 to establish a result required for our subsequent development.

Lemma 1: Fix any \( w \in \mathbb{C}^L \) and \( a, b \) such that \( \mathcal{G}(w, a, b) \) is a frame for \( \mathbb{C}^L \), with \( M = L/b \). Then for any integral \( M' \geq \text{len}(w) \), the Gabor system \( \mathcal{G}(w, a, L/M') \) is also a frame for
III. SUPERPOSITION WINDOWS

Having outlined the connections between Gabor systems and the short-time Fourier transform, we now introduce the central ingredient of our signal-adaptive time-frequency analysis framework—the superposition window construction, illustrated in Figure 1(a).

Definition 6 (Superposition Window): Fix a real, nonnegative window \( w \) on \( \mathcal{C}^L \) and an integer \( a = L/N \), along with some \( r \in \mathbb{Z}_N \). We then define the superposition window \( w_r \) to be a linear sum of \( r + 1 \) adjacent translates of \( w[t] \) as follows:

\[
w_r[t] \triangleq \sum_{n=0}^{r} T_{na} w[t], \quad r \in \mathbb{Z}_N.
\]

(8)

Remark 5 (Fourier Transform Support): Let \( \hat{w} \) denote the (discrete) Fourier transform of \( w \in \mathcal{C}^L \). Linearity of \( \mathcal{G} \) implies that the support of \( \hat{w}_r \) is contained within that of \( \hat{w} \), as supp(\( \hat{w}_r \)) = supp(\( \sum_{n=0}^{r} e^{-2\pi i n a / L} \hat{w} \)) \( \subseteq \) supp(\( \hat{w} \)).

Remark 6 (Fourier Transform Decay): As \( r \) increases, it is clear that \( w_r \) can become more like a rectangular window (see, e.g., Fig. 1(a)); this effect is illustrated in Fig. 2 for the case of Hamming superposition windows. Consequently, the main lobe width of \( w_r \) shrinks, leading to improved frequency resolution relative to \( \tilde{w}_0 \); this main lobe resolution, however, comes at the expense of decreasing sidelobe attenuation. Spectral leakage—a function of the window smoothness—remains superior, as does overall spectral decay for small \( r \).

Our subsequent construction of superposition frames employs sets of modulated superposition windows, and to this end we establish the following two energy “conservation” properties, proved in the appendix.

Lemma 2 (Localized Parseval Property): Fix any \( w \in \mathcal{C}^L \) and an integer \( M = L/b \geq \text{len}(w) \). Then

\[
\forall x \in \mathcal{C}^L, \quad \sum_{m=0}^{M-1} |\langle x, \mathcal{M}_{mb} w \rangle|^2 = M \sum_{t=0}^{L-1} |x[t]|^2 |w[t]|^2.
\]

(9)

Lemma 3 (Superadditivity of Superposition Energy): Let real, nonnegative superposition windows \( w_p \) and \( w_q \) be derived from a Gabor system \( \mathcal{G}(w, a, b) \) on \( \mathcal{C}^L \), and merge them to obtain a new superposition window \( w_p + w_q' \) = \( w_p + T_{(p+1)a} w_q \). Then, if and only if \( M = L/b \geq \text{len}(w_p + w_q') \), the following holds for every \( M_0 = L/b_0 \in \{\max(\text{len}(w_p), \text{len}(w_q)), \ldots, M\} \):

\[
M_0 - 1 \sum_{m=0}^{M_0-1} |\langle x, \mathcal{M}_{mb} w_p \rangle|^2 + |\langle x, \mathcal{M}_{mb} w_q \rangle|^2 \\
\leq \sum_{m=0}^{M-1} |\langle x, \mathcal{M}_{mb} (w_p + w_q') \rangle|^2, \quad \forall x \in \mathcal{C}^L.
\]

(10)
adapting a Gabor frame through the superposition construction preserves the original Gabor lower frame bound—an important consideration for numerical stability.

IV. CONSTRUCTION OF SUPERPOSITION SYSTEMS

We now describe how to employ the superposition windows of Section III above to create a signal-adaptive analysis framework. Let $\mathcal{G}(w, a, b)$ represent a Gabor system, which induces a short-time Fourier transform on $\mathcal{C}^L$ according to Definition 1. Beginning with $\mathcal{G}(w, a, b)$, we then form a signal-dependent, variable-resolution STFT by using the superposition sum of (8) to adaptively merge neighboring translates from the set $\{T_{n}w, n \in \mathbb{Z}_N\}$. Later we will demonstrate how this signal-adaptive analysis can be coupled with a variety of different algorithms; we begin, however, by studying the general set of superposition systems independently of any algorithmic construction. To this end, we introduce the notion of ordered partition functions as a means of indexing arbitrary sets of superposition windows, and then extend these to yield a full time-frequency analysis.

A. Ordered Partition Functions

Observe that exactly $2^{N-1}$ distinct sets of variable-length superposition windows may be derived by merging window translates from a given Gabor system $\mathcal{G}(w, a, \cdot)$. As a means of indexing these sets, the following “stick-breaking” analogy is helpful. Consider a “stick” composed of $N$ ordered, unit-length pieces, representing elements of the set $\{T_{n}w, n \in \mathbb{Z}_N\}$. Merging adjacent windows in this set can be thought of as fusing neighboring pieces of the stick. Each stick partition thus induces an ordered partition of the set $\{1, 2, \ldots, N\}$, with each piece uniquely identified by an initial index and length, and we may formalize this analogy as follows.

**Definition 7 (Ordered Partition Functions):** We call any $\tilde{I} : \mathbb{Z}_N \times \mathbb{Z}_N \rightarrow \{0, 1\}$ an ordered partition function if it is not identically zero, and satisfies the following three properties:

1. Each piece of the stick is distinct:
   $$\tilde{I}[n, r] = 1 \Rightarrow \tilde{I}[n, r'] = 0 \ \forall r' \neq r, r' \in \mathbb{Z}_N.$$

2. The length of each piece is denoted by $r + 1$:
   $$\tilde{I}[n, r] = 1 \Rightarrow \tilde{I}[n', r'] = 0 \ \text{on} \ \{n+1, \ldots, n+r\} \times \mathbb{Z}_N.$$

3. All pieces of the stick are accounted for:
   $$\tilde{I}[n, r] = 1 \Rightarrow \tilde{I}[n+r, r'] = 1 \ \text{for exactly one} \ r' \in \mathbb{Z}_N.$$

**Example 1 (N-Part Partition):** The ordered partition function associated to the top panel of Fig. (1b) is

$$\tilde{I}[\cdot, r] \triangleq \begin{cases} 1 & \text{if } r = 0, \\ 0 & \text{otherwise.} \end{cases}$$

This clearly recovers the window translates of any Gabor system $\mathcal{G}(w, a, \cdot)$. Note that in accordance with (11), we have that

$$\sum_{n=0}^{N-1} \sum_{r=0}^{N-1} \tilde{I}[n, r] = 1 \Rightarrow \sum_{n=0}^{N-1} \tilde{I}[n, 0] = N = 8.$$  

**Example 2 ((N-3)-Part Partition):** The ordered partition function associated to the bottom panel of Fig. (1b) is

$$\tilde{I}[n, r] \triangleq \begin{cases} 1 & \text{not merged,} \\ 0 & \text{merged.} \end{cases}$$

Note again that in accordance with (11), we have that

$$\sum_{n=0}^{N-1} \sum_{r=0}^{N-1} \tilde{I}[n, r] = 3 + 2 + 3 = 8.$$

B. Superposition Systems

We now employ the above construction to arrive at a variable-resolution time-frequency analysis via superposition windows. To this end, let the set $\mathcal{F}$ be defined as a function of any Gabor system $\mathcal{G}(w, a, b)$ on $\mathcal{C}^L$ as follows:

$$\mathcal{F} \triangleq \bigcup_{r \in \mathbb{Z}_N} \mathcal{G}(w_r, a, b_L),$$

where the frequency lattice $b_L \mathbb{Z}$ encompasses all possible Gabor systems on $\mathcal{C}^L$ for a fixed choice of integral $M$:

$$b_L \mathbb{Z} ; M_L \triangleq \text{lcm}(\{1, 2, \ldots, \text{max}(L, M)\}), \ b_L \triangleq L / M_L.$$  

Elements of $\mathcal{F}$ may then be defined in analogy to (2) as

$$\phi_{m, n, r} \triangleq M_{mb_L} T_{na} w_r = M_{mb_L} (\sum_{n'=0}^{N-1} T_{(n'+n)a} w),$$

and in turn give rise to superposition systems, defined as appropriately chosen subsets of $\mathcal{F}$. 

**Definition 8 (Superposition Systems and Admissibility):** Fix an ordered partition function $I[n, r]$ and a function $M[n, r] : \mathbb{Z}_N \times \mathbb{Z}_N \rightarrow \mathbb{Z}_{M_L}$. We call any $I[m, n, r] : \mathbb{Z}_{M_L} \times \mathbb{Z}_N \times \mathbb{Z}_N \rightarrow \{0, 1\}$ an admissible selection function on $\mathcal{F} = \bigcup_r \mathcal{G}(w_r, a, b_L)$ if it satisfies the following two properties:

$$I[0, n, r] = I[n, r] \ \forall n, r \in \mathbb{Z}_N \times \mathbb{Z}_N,$$

$$I[0, n, r] = 1 \Rightarrow I[M[n, r] m, n, r] = 1, \ \forall m \in \mathbb{Z}_{M[n, r]}.$$  

Furthermore, we call the induced set of elements a superposition system $\mathcal{F}(I)$:

$$\phi_{m, n, r} \in \mathcal{F}(I) \Leftrightarrow I[m, n, r] = 1.$$  

It follows from (12) that the first $M[n, r]$ modulates of each selected superposition window are included in $\mathcal{F}(I)$, and thus we later suppress the dependence of $I$ on frequency bin index $m$ when possible, by abbreviating $I[\cdot, n, r]$ as $I[n, r]$.  

$$I'[\cdot, n, r]$$
V. Superposition Frames: Main Results

Starting from a Gabor system $\mathcal{G}(w,a,b)$, we see that any superposition system $\mathcal{F}(I) \subset \cup_r \mathcal{G}(w_r,a,b_L)$ effectively yields a “variable-resolution” subsampled short-time Fourier transform, defined for all $m \in \mathbb{Z}_{ML}$ and $n, r \in \mathbb{Z}_N$ as

$$X[m,n,r] \triangleq \begin{cases} \langle x, \phi_{m,n,r} \rangle & \text{if } \phi_{m,n,r} \in \mathcal{F}(I), \\ 0 & \text{otherwise}. \end{cases} \quad (13)$$

Consequently, we now establish conditions under which superposition systems $\mathcal{F}(I)$ form frames for $\mathbb{C}^L$, in analogy to the relation between a Gabor frame and its corresponding fixed-resolution short-time Fourier transform.

A. General Case: Sufficiency

To begin our analysis, consider the case of an admissible selection function $I[m,n,r]$ for which $M[n,r] = M_g$ for all $n, r \in \mathbb{Z}_N$, corresponding to the notion of a global frequency lattice of arbitrary resolution: $b_g \mathbb{Z}$ with $b_g = L/M_g$. Our first result, proved in the appendix, ensures that the induced superposition system $\mathcal{F}(I)$ is a frame for $\mathbb{C}^L$ if the following test condition holds.

Theorem 1 (Sufficiency Condition, Superposition Frames): Fix a Gabor system $\mathcal{G}(w,a,\cdot)$ on $\mathbb{C}^L$, with $N = L/a$, $w$ real and nonnegative, and define for $s, t \in \mathbb{Z}_L$, $n, r \in \mathbb{Z}_N$, the term

$$\beta_{nr}(s,t) \triangleq \frac{w_s}{w_t} [t-na] [t-na-s].$$

Let $I[n,r]$ be any admissible selection function for which $M[n,r] = M_g$ for some $M_g \in \{1,2,\ldots,L\}$. Then, for index term $k \in \{\{t-(L-1)/M_g\}, \ldots, \{t/M_g\}\}$, the condition

$$\forall t \in \mathbb{Z}_L, \sum_{n=0}^{N-1} \sum_{r=0}^{N-1} I[n,r] (\beta_{nr}(0,t) - \sum_{k \neq 0} \beta_{nr}(k M_g, t)) > 0 \quad (14)$$

is sufficient to guarantee that the superposition system $\mathcal{F}(I) \subset \cup_r \mathcal{G}(w_r,a,b_L)$ is a frame for $\mathbb{C}^L$.

Satisfying the criterion of (14) implies that the underlying frame operator is strictly diagonally dominant—a sufficient condition for strict positive definiteness. This is a popular criterion in the literature (see, e.g., [32, Corollary 6], [33, Theorem 8.4.4]) and, as can be seen from (14), takes a particularly simple form in the superposition setting.

B. Superposition Frames and Frame Bounds

In Theorem 1 above, we considered a general class of superposition frames associated with an arbitrary frequency lattice $b_g \mathbb{Z}$ in Theorems 2 and 3 below, we study two distinct classes of superposition systems using non-uniform (local) and uniform (global) modulation structures defined as follows.

Definition 9 (Admissible Selection Functions $I^l$ and $I^g$): Fix a Gabor system $\mathcal{G}(w,a,L/M)$, associate to it any ordered partition function $I[r,\cdot]$, and define

$$M_r \triangleq \max \{\text{len}(w_r), M\}; \quad b_r \triangleq L/M_r; \quad (15)$$

$$M_g \triangleq \max_{r: I[r,\cdot] = 1} M_r; \quad b_g \triangleq L/M_g. \quad (16)$$

These quantities induce, via $M[n,r] = M[\cdot,r] = M_r$ or $M[n,r] = M_g$ constant, respective classes of local and global admissible selection functions $I^l[m,n,r]$ and $I^g[m,n,r]$.

We now show that superposition systems $\mathcal{F}(I^g)$ and $\mathcal{F}(I^l)$ are frames for $\mathbb{C}^L$. Later, we will verify that such frames admit diagonal frame operators. This special structure leads not only to fast reconstruction algorithms, but also to the preservation of lower frame bounds.

Theorem 2 (Local and Global Superposition Frames): Let $\mathcal{G}(w,a,b)$ be a Gabor frame for $\mathbb{C}^L$, with $w$ real and nonnegative. Then for any choice of admissible selection functions $I^l$ and $I^g$, the local and global superposition systems $\mathcal{F}(I^g)$ and $\mathcal{F}(I^l)$ are also frames for $\mathbb{C}^L$.

Proof: As our finite-dimensional setting implies the existence of an upper frame bound for any admissible $I$, only the existence of a lower frame bound need be established. The proof proceeds via Lemma 3 and an iterative argument.

To begin, consider the admissible selection function $I^g[m,n,r]$ induced by an $(N-1)$-part ordered partition, which is associated to the event that no merging of windows in $\mathcal{G}(w,a,b)$ occurs, and hence $\mathcal{F}(I^g) = \mathcal{G}(w,a,b_g)$, with $b_g = L/M_g$ and $M_g = \max\{\text{len}(w), M\}$ according to (16). Lemma 3 then ensures that $\mathcal{F}(I^g) = \mathcal{F}(I^l)$ is a frame for $\mathbb{C}^L$, with maximal lower frame bound $M_g = \min_{x \in \mathbb{C}^L} \sum_{n=0}^{N-1} |\langle x, T_n w \rangle|^2 > 0$.

Next consider any admissible selection function $I^l[m,n,r]$ induced by an $(N-1)$-part ordered partition, corresponding to the case that exactly one pair of windows $w_0 \equiv w_0$ from the initial Gabor frame $\mathcal{G}(w,a,b)$ is merged via the superposition sum (8). In this case, there exists one $n^* \in \mathbb{Z}_N$ such that $I^l(0,n^*,1) = 1$, and so (12) implies that $\mathcal{F}(I^l)$ contains the elements $\{T_{n^*} w_0 \mid m \in \mathbb{Z}_{M_g}\}$. Each of these elements can in turn be decomposed into the following sum:

$$M_{mb} T_{n^*} w_0 = M_{mb} T_{n^*} w_0 + M_{mb} T_{(n^*+1)a} w_0. \quad (17)$$

Since (12) implies $M_g \geq \text{len}(w_1) = \text{len}(w_0 + T_a w_0)$, we obtain by (17) and the superadditivity property of Lemma 3

$$\sum_{m=0}^{M_{mb} - 1} |\langle x, M_{mb} T_{n^*} w_0 \rangle|^2 + \sum_{m=0}^{M_{mb} - 1} |\langle x, M_{mb} T_{(n^*+1)a} w_0 \rangle|^2 \quad \forall x \in \mathbb{C}^L. \quad (18)$$

Next, noting that by the decomposition of (17), we have

$$\mathcal{F}(I^l) = \mathcal{G}(w,a,b_g) \cup \{M_{mb} T_{n^*} w_0\} \cup \{M_{mb} T_{(n^*+1)a} w_0\}, \quad (19)$$

we see that (13) and (19) together imply that for all $x \in \mathbb{C}^L$,

$$\sum_{\phi \in \mathcal{F}(I^l)} |\langle x, \phi_{m,n,r} \rangle|^2 \geq \sum_{\phi \in \mathcal{G}(w,a,b_g)} |\langle x, \phi_{m,n} \rangle|^2. \quad (20)$$

Since $\mathcal{G}(w,a,b_g) = \mathcal{F}(I^g)$ is a frame, the existence of a lower frame bound for $\mathcal{F}(I^l)$ is guaranteed by (20).
Now consider the general case in which $\mathcal{F}(I^k)$ contains multiple merge functions. Since our construction ensures that all admissible selection functions can be obtained by iterative partitioning in the manner above, we can always recover the inequality of (20) for $\mathcal{F}(I^k)$ and any $b_g$ according to (16), by linearity of superposition and repeated application of Lemma 3.

A similar iterative argument holds for $\mathcal{F}(I^l)$, with $b_g$ replaced by $b_0$ from (15). In place of (19) we obtain

$$\mathcal{F}(I^l) = (\mathcal{G}(w, a, b_0) \cup \{M_{mb0} \mathcal{T}_{n,a} w_0\}) \setminus \{(M_{mb0} \mathcal{T}_{n,a} w_0) \cup \{M_{mb0} \mathcal{T}_{(n+1,a)} w_0\}\},$$ (21)

with $b_l = L/M_1$ according to (15). Note that

$$M_1 \geq \text{len}(w_1) = \text{len}(w_0 + \mathcal{T}_a w_0),$$

$$M_0 \geq \text{len}(w_0),$$ and $M_1 \geq M_0$.

Thus, we may apply Lemma 3 to the latter three terms of (21), yielding the required result for $\mathcal{F}(I^l)$: for all $x \in \mathbb{C}^L$,

$$\sum_{\phi \in \mathcal{F}(I^l)} |\langle x, \phi_{m,n,r} \rangle|^2 \geq \sum_{\phi \in \mathcal{G}(w,a,b_0)} |\langle x, \phi_{m,n,r} \rangle|^2.$$

As above, the proof for general $\mathcal{F}(I^l)$ then follows.

Thus we see that for every Gabor system on $\mathbb{C}^L$ and any associated ordered partition function, setting $M[n,r]$ in accordance with $M_1$ or $M_2$ will yield local or global superposition frames. Moreover, as we detail later, the iterative arguments employed above suggest precise algorithmic constructions.

We now proceed to establish the important property that, for all local and global $I[n,r]$ of Definition 9, superposition frames preserve lower frame bounds, thus ensuring numerical stability of the resultant representation. The following result also formulates the corresponding minimax-optimal superposition frame bounds.

**Theorem 3 (Superposition Frame Bound Properties):** Let $\mathcal{G}(w, a, b)$ be a frame for $\mathbb{C}^L$, with associated maximal lower frame bound $A > 0$. Then for any admissible $I^k$ or $I^l$:

1) The quantity $A$ remains a valid lower frame bound for both $\mathcal{F}(I^k)$ and $\mathcal{F}(I^l)$.

2) The **minimum** maximal lower superposition frame bound over all admissible $I^k$ and $I^l$ is

$$A_{\text{opt}} = \frac{L}{b_{\text{max}}} \min_{t \in \mathbb{Z}_L} \left(\sum_{n=0}^{N-1} |\mathcal{T}_n w[t]|^2\right),$$

with $b_{\text{max}} = \min(b, L/\text{len}(w))$. It is attained in the absence of merging: $\mathcal{F}(I^k) = \mathcal{F}(I^l) = \mathcal{G}(w, a, b_{\text{max}})$.

3) The **maximum** minimal upper superposition frame bound over all admissible $I^k$ and $I^l$ is

$$B_{\text{opt}} = \frac{L}{b_{\text{min}}} \max_{t \in \mathbb{Z}_L} \left(\sum_{n=0}^{N-1} |\mathcal{T}_n w[t]|^2\right),$$

with $b_{\text{min}} = \min(b, 1)$. It is attained when all translates of $w$ have been merged: $\mathcal{F}(I^k) = \mathcal{F}(I^l) = \mathcal{G}(w_{N-1}, L, b_{\text{min}})$, with $w_{N-1} = \sum_{n=0}^{N-1} \mathcal{T}_n w[t]$.

**Proof:** Let $S$ be the frame operator associated to $\mathcal{G}(w, a, b)$, with smallest eigenvalue $A$, and observe that $\min_{t \in \mathbb{Z}_L} S[t,t] \geq A$ by the Schur-Horn convexity theorem. Now, for any admissible $I^k$ or $I^l$, the proof of Theorem 2 shows that the maximal lower frame bound of $\mathcal{F}(I^k)$ or $\mathcal{F}(I^l)$ is bounded from below by that of some $\mathcal{G}(w, a, b')$, with $b'$ denoting respectively $b_g$ or $b_0$. Therefore, (1) will follow if we can show the maximal lower frame bound of $\mathcal{G}(w, a, b')$ to be no less than $\min_{t \in \mathbb{Z}_L} S[t,t]$. To do so, note that Lemma 1 implies that the frame operator $S'$ of $\mathcal{G}(w, a, b')$ is diagonal, with smallest eigenvalue $\min_{t \in \mathbb{Z}_L} S'[t,t] = (M'/M) \min_{t \in \mathbb{Z}_L} S[t,t]$; (15) and (16) then yield $M' \geq M$.

Next recall that superposition frames are induced from a Gabor frame $\mathcal{G}(w, a, b)$ by merging $n \in \mathbb{Z}_N$ neighboring window translates $\mathcal{T}_n w$. By the above argument, $\min_{t \in \mathbb{Z}_L} S'[t,t]$ is itself a maximal lower frame bound for the case $\mathcal{G}(w, a, b')$ attained whenever no window translates are merged; likewise, the merging of all translates yields $\mathcal{G}(w_{N-1}, L, b''')$ for some unique $b'''$, with minimal upper frame bound $\max_{t \in \mathbb{Z}_L} S'''[t,t]$.

To show that these cases are in fact extremal as claimed, we appeal to the same iterative argument used to prove Theorem 2. There, the superadditivity property of Lemma 3 was invoked to show that for any admissible $I^k$ or $I^l$, merging superposition windows cannot decrease the overall energy of the resultant frame coefficients. Thus, the case of $\mathcal{G}(w, a, b')$ considered above represents attainment of the minimum maximal lower superposition frame bound, and moreover $M' = \max(M, \text{len}(w))$ via (15) and (16). Likewise, $\mathcal{G}(w_{N-1}, L, b''')$ yields the maximum minimal upper superposition frame bound, with $M'' = \max(M, \text{len}(w_{N-1}))$. Lemma 2 then establishes the bound directly:

$$\sum_{\phi \in \mathcal{G}(w_{N-1}, L, b''')} |\langle x, \phi_{m,n,r} \rangle|^2 = \sum_{m=0}^{M'} |\langle x, M_{mb', w_{N-1}} \rangle|^2 = M'' \sum_{t=0}^{L-1} |x[t]|^2 w_{N-1}[t]^2 \leq \|x\|^2 M'' \max_{t \in \mathbb{Z}_L} |w_{N-1}[t]|^2,$$

and the proof is completed by noting that as $\mathcal{G}(w, a, b)$ is assumed a Gabor frame for $\mathbb{C}^L$, the covering condition of Remark 2 implies that $|\text{supp}(w_{N-1})| = \text{len}(w_{N-1}) = L$, and hence $M'' = \max(M, L)$ as claimed.

**VI. Fast Reconstruction via Superposition Frames**

We now show how the special structure of our superposition construction gives rise to a number of efficient reconstruction procedures. For any signal of interest $x \in \mathbb{C}^L$, the superposition frame analysis coefficients $X[m,n,r] = I[m,n,r] x[\phi_{m,n,r}]$ can be computed via fast Fourier transform (FFT) once an admissible selection function has been specified. Superposition frames also enable fast (FFT-based) reconstruction from the corresponding analysis coefficients, in contrast to the general case of $O(L^3)$ complexity for frame-based reconstruction via inversion of the frame operator.

We first provide a fast constant-overlap-add reconstruction method, which obviates the need for canonical dual frames. We next show that reconstruction via the canonical dual can also proceed by way of a pointwise modification of each superposition window $\phi_{0,0,r}$, followed by the application of FFTs, as in the case of general nonstationary Gabor frames [31]. Third, we show that in settings reminiscent of lapped orthogonal transforms, calculation of canonical dual windows is possible.
independently of any $I^g$—in contrast to the typical signal-adaptive setting, where the structure of the frame operator is a function of the instantiated signal adaptation. Last, we compare the computational complexity of these procedures.

A. Reconstruction via the Constant Overlap-Add Method

The classical “overlap-add” approach to signal reconstruction from short-time Fourier coefficients proceeds as follows [35]. Recall the covering condition of (6) which is necessary for a Gabor system $\mathcal{G}(w, a, b)$ to form a frame for $\mathbb{C}^L$ and also sufficient if $M = L/b \geq \text{len}(w)$. Clearly, this covering condition holds if translates $\{\mathcal{T}_{na}w : n \in \mathbb{Z}_N\}$ form a partition of unity on $\mathbb{C}^L$ (see, e.g., the top panel of Fig. 1(b)), and to this end we obtain the following definition, long popular in the signal processing literature.

Definition 10 (Constant Overlap-Add Window Constraint): Fix a Gabor system $\mathcal{G}(w, a, \cdot)$ on $\mathbb{C}^L$. Then, noting the discrete Fourier transform evaluation $\hat{w}[0] = \sum_{t=0}^{N-1} w[t]$, the window $w$ is said to satisfy the constant overlap-add constraint if

$$\forall t \in \mathbb{Z}_L, \sum_{n=0}^{N-1} w[t-na] = \frac{\hat{w}[0]}{a}. \quad (22)$$

To clarify the role of this overlap-add constraint in fast reconstruction, consider a Gabor frame $\mathcal{G}(w, a, b)$ on $\mathbb{C}^L$ for which $w$ satisfies (22), with $M = L/b$ chosen such that $M \geq \text{len}(w)$. The associated short-time analysis coefficients $\{X[m, n] : m \in \mathbb{Z}_M, n \in \mathbb{Z}_N\}$ are obtained as inner products of any $x \in \mathbb{C}^L$ according to (1), and it is easy to show that

$$x[t] = \frac{a}{\hat{w}[0]} \sum_{n=0}^{N-1} \left( \frac{1}{M} \sum_{m=0}^{M-1} X[m, n]e^{2\pi imbt/L} \right). \quad (23)$$

The constraint thus admits a reconstruction procedure based on the overlapping additions of a sequence of discrete Fourier transforms on $\mathbb{C}^M$.

Remark 7 (Superposition Windows Preserve Overlap-Add): By their linear construction, superposition windows $w_r$ preserve the constant overlap-add constraint of Definition 10 for any Gabor system $\mathcal{G}(w_r, a_r)$. To see this, note that the discrete Poisson summation formula on $\mathbb{C}^L$, for $N = L/a$, is given by $\sum_{n=0}^{N-1} \hat{x}[t-na] = a^{-1} \sum_{k=0}^{a-1} e^{2\pi i kNt/L}$. Applying this expression to $\{\mathcal{T}_{na}w : n \in \mathbb{Z}_N\}$ yields the relation

$$\sum_{n=0}^{N-1} w[t-na] = \frac{1}{a} \sum_{k=0}^{a-1} e^{2\pi i kNt/L} \hat{w}[kN],$$

and it follows that the constraint of (22) holds (for a given time lattice constant $a$) if the Fourier transform $\hat{w}$ satisfies

$$\hat{w}[kN] = 0, \forall k \in \{1, \ldots, a - 1\}.$$
Perfect reconstruction then follows from the generalized overlap-add constraint of (24), as
\[
\frac{a}{\hat{w}[0]} \sum_{n=0}^{N-1} \sum_{r=0}^{N-1} I^g[n, r]T_{na}w_r[t]x[t] = \frac{\hat{a}}{\hat{w}[0]} x[t] = x[t].
\]

For the case of a local selection function \(I^l\), note that the generalized overlap-add constraint of (24) is still implied by (22), since the argument for the case of admissible \(I^g\) holds independently of the modulation structure employed. Consequently, by substituting \(M_g, b_g\) for \(M_f, b_f\) and noting that \(M_f \geq \text{len}(\phi_{0, n, r})\) by (15) for each \(\phi_{0, n, r} \in \mathcal{F}(I^l)\), we see that the result of (25) also holds for all \(\mathcal{F}(I^l)\).

\[\square\]

**B. Reconstruction via Canonical Dual Superposition Frames**

We next develop the reconstruction properties of our superposition families in a frame-theoretic context, noting that they qualify as “painless nonorthogonal expansions” [36], and that the development below also holds for more general nonstationary Gabor frames [31]. In analogy to Definition 4, we associate a superposition frame operator \(S_I : \mathbb{C}^L \rightarrow \mathbb{C}^{L/a}\) through its action on any \(\mathbf{x} \in \mathbb{C}^L\) as \(S_I \mathbf{x} = \sum_{\mathbf{x} \in \mathcal{F}(I)} (x, \phi_{m, n, r})\phi_{m, n, r}\).

**Definition 11 (Superposition Frame Operator; Walnut Form):**

We thus denote by \(S_I\) and \(S_{I^l}\) the corresponding superposition frame operators, and observe the following consequence of the Walnut representations of superposition frame operators:

\[
S_{I^l}[t, t'] \triangleq M_g I^g_{M_g \setminus (t, t')} [t - t'] \sum_{n, r=0}^{N-1} I^g[n, r]T_{na}w_r[t]T_{na}w_r[t'],
\]

\[
S_{I^l}[t, t'] \triangleq I^l[n, r]M_g \cdot I^g_{M_g \setminus (t, t')} [t - t'] T_{na}w_r[t]T_{na}w_r[t'],
\]

\[
S_{I^l}[t, t'] \triangleq \sum_{n, r=0}^{N-1} I^l[n, r]M_g \cdot I^g_{M_g \setminus (t, t')} [t - t'] T_{na}w_r[t]T_{na}w_r[t'],
\]

\[
S_{I^l}[t, t'] \triangleq S_I^T \phi_{m, n, r}.
\]

Theorem 2 implies that any Gabor frame \(\mathcal{G}(w, a, b)\) and admissible \(I^g\) or \(I^l\) together give rise to a superposition frame, and hence the corresponding superposition frame operators are of full rank. Thus, to each global superposition frame \(\mathcal{F}(I^g) = \{\phi_{m, n, r}\}\) corresponds a unique canonical superposition frame \(\{\phi_{m, n, r}\}\), whose elements are obtained in turn as \(\phi_{m, n, r} = \phi_{m, n, r} \sim \phi_{m, n, r}\).

Accordingly, when \(S_I\) is diagonal we may index elements of \(\{\phi_{m, n, r}\}\) by the same admissible \(I^g\), and we obtain the following reconstruction property:

\[
\forall x \in \mathbb{C}^L, t \in \mathbb{Z}_L, x[t] = \sum_{m, n, r, I^g[m, n, r]=1} \langle x, \phi_{m, n, r} \rangle \hat{\phi}_{m, n, r}[t],
\]

with the above also holding for local \(I^l\) by Theorem 2.

We thus denote by \(\mathcal{F}(I^g)\) or \(\mathcal{F}(I^l)\) the corresponding dual frames, and observe the following consequence of the Walnut representation of Definition 11.

**Theorem 5 (Fast Inversion via Canonical Dual):** For any \(\mathcal{F}(I^g)\), \(\mathcal{F}(I^l)\) derived from a Gabor frame \(\mathcal{G}(w, a, b)\) on \(\mathbb{C}^L\), the corresponding operators \(S_I\) and \(S_{I^l}\) are diagonal, and each canonical dual frame element takes the form

\[
\sim \phi_{m, n, r}[t] = \frac{M_{mb_a}T_{na}w_r[t]}{M_g \sum_{n'=0}^{N-1} \sum_{r'=0}^{N-1} I^g[n', r'] \| T_{na}w_r[t] \|^2},
\]

for \(\mathcal{F}(I^g)\), and similarly for \(\mathcal{F}(I^l)\) with respect to each \(M_r\).

Note that the corresponding formula for the nonstationary Gabor frames of [31] in the diagonal case is similar to (27). However, in the superposition frame setting, the constraints on the window structure not only preserve lower frame bounds and yield fast inversion via the constant overlap-add method, but also enable signal-independent evaluation of the canonical dual in certain cases, as we now show.

**C. Adaptive Lapped Superposition Frames**

Reconstruction via the canonical dual \(\mathcal{F}(I^g)\) according to (27) requires knowledge of the admissible selection function \(I^g[n, r]\) corresponding to a given signal adaptation. This stands in contrast not only to the usual Gabor setting, wherein the form of the canonical dual frame can be obtained immediately, but also to the constant overlap-add approach described in Section VI-A which avoids compution of the canonical dual entirely. However, by coupling our superposition construction with the following neighbor overlap condition, we are able to compute \(\mathcal{F}(I^g)\) prior to adaptation—that is, without knowledge of which ordered partition function will be used in subsequent signal analysis.

**Definition 12 (Neighbor Overlap Condition):** Let \(\mathcal{G}(w, a, \cdot)\) be a Gabor system on \(\mathbb{C}^L\), with \(N = L/a\). It is said to satisfy the neighbor overlap condition if, for all \(n, n' \in \mathbb{Z}_N\),

\[
\text{supp}(T_{na}w) \cap \text{supp}(T_{na}w) = \emptyset \quad \text{if} \quad |n - n'| > 1.
\]

Any admissible selection function preserves the neighbor overlap property, leading to the following notion of lapped superposition frames, whose properties we develop below.

**Definition 13 (Adaptive Lapped Superposition Frames):**

Let \(\mathcal{G}(w, a, \cdot)\) be a Gabor frame on \(\mathbb{C}^L\) that simultaneously satisfies the overlap-add constraint of (22) and the neighbor-overlap condition of (28). Then for any admissible \(I^g\), we call \(\mathcal{F}(I^g)\) an adaptive lapped superposition frame.

Note that the overlap-add constraint of (22) ensures a partition of unity by window translates, while the neighbor-overlap condition of (28) is also required in the case of lapped orthogonal transforms (see, e.g., [7]). While our construction retains the flavor of time-varying lapped transforms [37], [38], we emphasize that the resultant frames can avoid the lack of translation invariance inherent in the orthogonal setting, while still ensuring fast reconstruction.

We show below that if \(\mathcal{F}(I^g)\) is a lapped superposition frame derived from a Gabor frame \(\mathcal{G}(w, a, b)\), then its canonical dual frame elements may be pre-computed. This situation is illustrated in Fig. 5 where the support sets of a window \(w\), its canonical dual \(\hat{w}\), and their immediate neighbors are partitioned into subsets labeled \(L\), \(C\), and \(R\). Since \(\mathcal{F}(I^g)\) must inherit the neighbor-overlap condition from \(\mathcal{G}(w, a, b)\), it follows that whenever \(\mathcal{F}(I^g)\) admits a diagonal frame operator, the corresponding canonical dual windows of \(\mathcal{F}(I^g)\) are constant on the center set \(C\), as shown in Fig. 5. Moreover, the highlighted dual superposition window is pointwise equal to \(\hat{w}\) on the sets \(L\) and \(R\) (see bottom two panels of Fig. 5).
 frame families that all admit fast reconstruction. 

Thus enabling a variety of new adaptive, lapped superposition windows

\[
\{T_{n,a}w\} \text{ from a Gabor frame } \mathcal{G}(w, a) \text{ constructed from triangular windows with } 50\% \text{ overlap (top panel), shown with translates of its canonical dual window } \tilde{w} \text{ (second panel); remaining panels repeat this sequence for an induced superposition frame } \mathcal{F}(I^4). \text{ On the left and right sets } L \text{ and } R, \text{ the canonical dual windows of } \mathcal{F}(I^4) \text{ agree pointwise with those of } \mathcal{G}(\tilde{w}, a); \text{ on the center set } C, \text{ that of } \mathcal{F}(I^4) \text{ is constant.}
\]

To formalize this intuition, define for each \( r \) the sets

\[
\begin{align*}
L_r & \triangleq \text{supp}(w_r) \cap \text{supp}(T_{(r+1)a}w_r), \\
R_r & \triangleq \text{supp}(w_r) \cap \text{supp}(T_{(r+1)a}w_r), \\
C_r & \triangleq \text{supp}(w_r) \setminus (L_r \cup R_r),
\end{align*}
\]

and note that, for any global selection function \( I^8, \phi, n, r \in \mathcal{F}(I^4) \) and its dual \( S^{-1}_{I^8}\phi, n, r \in \mathcal{F}(I^4) \) are both supported exclusively on the set \( T_{n,a}(L_r \cup C_r \cup R_r) \). Then the following theorem, proved in the appendix, establishes our main result: the canonical dual frame of any \( \mathcal{F}(I^4) \) can be computed independently of any ordered partition function, and, therefore, can be computed \textit{prior} to observing any data.

**Theorem 6 (Canonical Duals of Adaptive Lapped Frames):** Let \( \mathcal{F}(I^8) \) arise from a Gabor frame \( \mathcal{G}(w, a, \cdot) \) for \( CL \) satisfying (22) and (28). Then every \( \phi_{m,n,r} \in \mathcal{F}(I^8) \) can be constructed by modulations \( M_{mb} \) and translations \( T_{na} \) of lapped windows corresponding to each \( r \) as follows:

\[
\tilde{\phi}_{0,0,r}[t] \triangleq \frac{1}{M_g} \begin{cases} 
\sum_{n=0}^{N-1} |w[n]|^2 & \text{if } t \in L_r, \\
\sum_{n=0}^{N-1} |w[n]|^2 & \text{if } t \in C_r, \\
\sum_{n=0}^{N-1} |w[n]|^2 & \text{if } t \in R_r.
\end{cases}
\]

Here the sets \( L_r, C_r, R_r \) are defined in (29), and we note that only the expression for \( C_r \) is to be employed when \( \mathcal{F}(I^4) \) is comprised entirely of modulations of \( w_{N-1} \).

We note that many popular Gabor systems \( \mathcal{G}(w, a, \cdot) \) satisfy the requirements of this theorem—including triangular, Hamming, and raised-cosine windows \( w \) at 50% overlap—thus enabling a variety of new adaptive, lapped superposition frame families that all admit fast reconstruction.

**D. Adaptive Dyadic Superposition Frames**

Here we construct a class of so-called \textit{dyadic} superposition frames that admit offline canonical dual construction even when the overlap-add constraint of (22) is \textit{not} satisfied. We base this construction on the notion of \textit{dyadic} ordered partition functions, which may be thought of as indexing binary trees.

**Definition 14 (Dyadic Admissible Selection Functions):**

Let the number of translates \( N \) of \( w \) in \( \mathcal{G}(w, a, b) \) be a power of two, and define the set \( \mathcal{H} \triangleq \{0, 1, \ldots, \log_2 N\} \). An ordered partition function \( \mathcal{I}[n, r] \) is \textit{dyadic} if it satisfies the conditions of Definition 7 and

\[
\mathcal{I}[n, r] = 1 \text{ only if } r = 2^h - 1 \text{ for some } h \in \mathcal{H}.
\]

We denote by \( \mathcal{I}[n, m, r] \) a \textit{dyadic} admissible selection function induced by \( \mathcal{I}[n, r] \) and a global frequency lattice constant \( b_g \) and \( M_g \text{ defined according to (16).}

Viewing the \( N \) translates of \( w \) in \( \mathcal{G}(w, a, b) \) as leaves of a binary tree of height \( \log_2 N \), a dyadic ordered partition function selects windows corresponding to some tree level \( h \).

**Definition 15 (Dyadic Gabor and Superposition Frames):**

Fix an initial Gabor frame \( \mathcal{G}(w, a, b) \), such that \( N = L/a \) is a power of two, and let \( h \in \mathcal{H} \) index height in a binary tree. Now restrict \( r \in \mathbb{Z}_n \) to the index set \( \mathcal{R} \triangleq \{2^h - 1\} \) and fix for each \( r \in \mathcal{R} \) a time lattice constant \( a_r \triangleq a(r+1) \), and a dyadic admissible selection function \( \mathcal{I}d \) with associated global frequency lattice constant \( b_g \). Then:

1. We define dyadic superposition Gabor frame \( G_{d,r}(b_g) \) for all \( r \in \mathcal{R} \), and their union \( G_{d,r}(b_g) \) as follows:

\[
G_{d,r}(b_g) \triangleq \mathcal{G}(w_r, a_r, b_g) \subseteq \mathcal{G}(w_r, a_r, b_L),
\]

\[
G_{d,r}(b_g) \triangleq \bigcup_{r \in \mathcal{R}} G_{d,r}(b_g) \subseteq \bigcup_{r \in \mathcal{R}} G_{d,r}(b_g)\subseteq \bigcup_{r \in \mathcal{R}} G_{d,r}(b_g).
\]

2. We call \( \mathcal{F}(I^d) \subseteq G_{d,r}(b_g) \) a dyadic superposition frame. The fact that \( \mathcal{F}(I^d) \) and every dyadic \( G_{d,r}(b_g) \) are frames for \( CL \) follows from the assumption that \( \mathcal{G}(w, a, b) \) is a frame, coupled with the result of Theorem 2.

An example of this construction is illustrated in Fig. 3. The left panel shows Hamming windows \( w \equiv w_0 \) at 50% overlap, along with the corresponding dyadically-indexed superposition windows \( w_1 \) and \( w_3 \), and an example dyadic superposition frame \( \mathcal{F}(I^4) \) at bottom right. The right panel shows canonical duals associated to each dyadic superposition Gabor frame \( G_{d,r}(\cdot) \), for \( r \in \{0, 1, 3\} \), along with the corresponding canonical dual \( \mathcal{F}(I^4) \) at bottom right. The fact that \( \mathcal{F}(I^4) \) contains elements from each of these individual dual Gabor frames \( \mathcal{G}_{d,r}(\cdot) \) is verified by the following theorem.

**Theorem 7 (Canonical Duals of Adaptive Dyadic Frames):**

Let a dyadic superposition frame \( \mathcal{F}(I^d) \subseteq G_{d,r}(b_g) \) arise from a Gabor frame \( \mathcal{G}(w, a, b) \) for \( CL \) satisfying the neighbor-overlap condition of (28). Then its canonical dual \( \mathcal{F}(I^d) \) can be computed by either path of the following commutative diagram, where \( S_{\mu} \) is the frame operator associated to \( \mathcal{F}(I^d) \), and \( S_{\mu}^t \) is that associated to each dyadic Gabor frame \( G_{d,r} \).

\[
\begin{align*}
G_{d,r}(b_g) \xrightarrow{(S_{\mu}^t)^{-1}} & \bigcup_{r \in \mathcal{R}} G_{d,r}(b_g) \\
\downarrow I^d & \xrightarrow{(S_{\mu})^{-1}} \mathcal{F}(I^d) \\
\mathcal{F}(I^d) & \xrightarrow{(S_{\mu}^t)^{-1}} \mathcal{F}(I^d)
\end{align*}
\]
Adaptation

7a

When reconstruction proceeds via inversion of the frame

an FFT-based approach requires

\( X \)

\( = \)

\( L \)

\( \) e.g., \([32]\), such schemes remain super-linear in

in some circumstances this complexity can be reduced (see,

diagonal frame operator lacking any special structure. While

that these algorithms all yield a diagonal frame operator,

the various reconstruction methods presented above. Recall

E. Computational Complexity

As in the case of lapped superposition frames in Section VI-C,

G

\( \widetilde{G}_d \)

and-right path in (30), and hence requires knowledge of

\( I^d \).

Recall that when no windows are merged, we recover

\( \widetilde{G}_d \) comprises

\( M_g = L/b_g \geq M \) modulations of \( N_g \leq N \) windows:

\( N_g = \sum_{n=0}^{N-1} \sum_{r=0}^{N-1} I^n[n, r] \) .\n
(31)

It follows that \( N_g M_g (\log_2 M_g + 1) \) complex multiplications

are required to compute the analysis coefficients \( X[m, n, r] \)

via \( N_g \) FFTs, followed by \( N_g M_g \log_2 M_g \) operations required

for the \( N_g \) inverse FFTs required for reconstruction.

If \( \mathcal{F}(I^g) \) arises from the special case described in Theo-

rem 6 then the elements of \( \mathcal{F}(I^g) \) can be pre-computed, and

only \( N_g M_g \) extra operations are necessary for multiplication

by the requisite canonical dual windows. In the general case,

elements of \( \mathcal{F}(I^g) \) must be computed directly via \( 27 \), as

a function of the chosen selection function \( I^g \). Appealing
to (31), we see that this computation can be accomplished

using another \( 2N_g M_g \) calculations, leading to a total of

\( N_g M_g (3 + \log_2 M_g) \) complex multiplications. This analysis

is also an upper bound for the worst-case complexity for any

\( \mathcal{F}(I^d) \) having the same ordered partition function as \( \mathcal{F}(I^g) \).

The various analysis and synthesis complexities discussed

above are summarized in Table 1. In practice, the complexity

of the signal adaptation procedure must also be taken into ac-

count. This complexity depends both on the method for search-
ing among ordered partition functions \( I[n, r] \), and the cost

function used to compare them. Since it is clearly infeasible
to compare all \( 2^{N-1} \) ordered partition functions by exhaustive

search, we next consider greedy and dynamic-programming-

based approaches below. In both cases, the complexity of

evaluating the associated cost functions increases with window

length, and therefore it is advisable in practice to set an upper

bound on the maximal number of window merges.

VII. SIGNAL ADAPTATION ALGORITHMS AND EXAMPLES

Signal-adaptive modification of an initial Gabor frame

\( \mathcal{G}(w, a, b) \) on \( \mathbb{C}^L \) via superposition can produce any one of the

possible \( 2^{N-1} \) superposition frames whose properties we char-

acterized in Sections VII and VII. We now detail two instances of

a broad class of signal adaptation algorithms, any of which can

be used to select a superposition frame for subsequent signal

analysis. We propose both greedy and dynamic programming

approaches in Section VII-A and illustrate their performance

with two brief examples in Section VII-B.

TABLE I

| Analysis Complexity | Synthesis Complexity | No Adaptation | Adaptation |
|---------------------|----------------------|---------------|------------|
| Overlap-Add         | Overlap-Add          | \( NM(1 + \log_2 M) \) | \( NM(1 + \log_2 M) \) |
| Canonical Dual      | Canonical Dual       | \( NM(1 + \log_2 M) \) | \( N_g M_g (1 + \log_2 M_g) \) |

Fig. 4. Repeated superposition merges of Hamming windows, following the structure of a binary tree, are shown in panels (a-c), along with the canonical dual windows associated to each corresponding Gabor frame (e-g). A dyadic superposition frame \( \mathcal{F}(I^d) \) can be formed from the selected unmodulated elements of (d), which in turn will admit the corresponding canonical dual windows shown in (h), computed according to \( 28 \).

Observe that computing the \( \mathcal{F}(I^d) \) via direct inversion of
its (diagonal) frame operator \( S_{I^d} \) corresponds to the down-
and-right path in (30), and hence requires knowledge of \( I^d \).
However, Theorem 7 implies that all elements in \( \mathcal{F}(I^g) \)
can be pre-computed by instead following the right-and-down path.
As in the case of lapped superposition frames in Section VI-C
the neighbor overlap condition of \( 28 \) plays a key role.

E. Computational Complexity

We now address the relative computational complexity of the
various reconstruction methods presented above. Recall
that these algorithms all yield a diagonal frame operator,
implying that only \( O(L) \) operations are required for its
inversion, compared to \( O(L^3) \) in the worst case of a non-
diagonal frame operator lacking any special structure. While
in some circumstances this complexity can be reduced (see,
e.g., [32]), such schemes remain super-linear in \( L \), rendering them impractical for use in applications with \( L \gg 1 \).

Recall that when no windows are merged, we recover a Gabor frame \( \mathcal{G}(w, a, b) \) containing \( NM \) elements, with \( Na = Mb = L \). If the associated frame operator is diagonal,
an FFT-based approach requires \( NM(1 + \log_2 M) \) complex
multiplications to compute the short-time analysis coefficients
\( X[m, n] \) from \( x \in \mathbb{C}^L \), with \( NM \) of these needed to obtain the
\( N \) individual short-time segments \( \{T_{na}w[t]\}x[t]\{T_{na}w[t]\} \).
When reconstruction proceeds via inversion of the frame
operator, then \( NM \log_2 M \) operations are required to compute
the necessary inverse FFTs, plus \( NM \) operations to multiply
each resultant segment by the appropriate dual window. In
the overlap-add setting, this window is the identity, and hence
these latter \( NM \) operations are avoided. An in-depth discus-
sion of the general (non-diagonal) case, including methods
based on the Zak transform, is given in [32] and [39].

In the case that windows are merged, note that a global
superposition frame \( \mathcal{F}(I^g) \) derived from \( \mathcal{G}(w, a, b) \) comprises
\( M_g = L/b_g \geq M \) modulations of \( N_g \leq N \) windows:

\( N_g = \sum_{n=0}^{N-1} \sum_{r=0}^{N-1} I^n[n, r] \).

(31)
A. Signal-Adaptive Superposition Frame Selection

The first of the two Gabor frame adaptation algorithms we describe is a simple greedy approach that can be implemented by “growing” a given window forward in time through successive attempts to merge it with its subsequent neighboring translations [1]. Whenever a proposed merge fails, the procedure resets and repeats, halting when the end of the data stream is reached (or, equivalently in our cyclic setting, when the initial window is once again encountered).

A decision whether or not to merge adjacent windows can be made based on any suitable cost function. As one example, we employ the time-frequency concentration measure appearing in the popular work of [3], [4] on adaptive optimal-kernel time-frequency representations. Specifically, consider a short-time segment \( x_{n,r}[t] = \{ T_{w,n} w[t] x[t] \}_{w \in \text{supp}(T_{w,n} w)[t]} \), and define its time-frequency concentration in the manner of [1]–[4]

\[
C(x_{n,r}) = \frac{\sum_{m=0}^{M_r-1} \left| \langle x, M_{mb} T_{w,n} w \rangle \right|^4}{\left( \sum_{m=0}^{M_r-1} \left| \langle x, M_{mb} T_{w,n} w \rangle \right|^2 \right)^2},
\]

with \( M_r, b_r \) defined via (15). This ratio of powers of norms of short-time Fourier coefficients is suggestive of an “empirical spectral kurtosis,” and has also been used in minimum entropy deconvolution [40]; other choices are also possible [14].

As shown in [3], maximizing (32) favors short-time segments that concentrate local signal energy within the smallest regions of the time-frequency plane. Indeed, below we obtain similar results on an example akin to the one employed in [3]: the resultant superposition frames comprise shorter windows near time-localized transients, and longer windows near oscillatory signal portions. The resulting procedure requires \( O(N) \) iterations and is summarized in Algorithm 1.

Algorithm 1 Adaptation via Greedy Selection [1]

**Initialization**
- Fix input data \( x \in \mathbb{C}^L \) and a Gabor frame \( \mathcal{G}(w, a, b) \)
- Set \( (p, n_p) = (0, 0) \) and initialize \( \tilde{I}(n, r) \) to be the \( N \)-part ordered partition function of Example 1

**Greedy Selection:** For \( n = 0, 1, \ldots, N - 1 \),
- Compute a merged window \( T_{n_p,w} u_{p+1} = T_{n_p,w} u_p + T_{n_a,w} \)
  \( C(x_{n_p,p+1}), C(x_{n,p}), \) and \( C(x_{n,0}) \) via (32)
  \- If \( C(x_{n_p,p+1}) \leq \max(C(x_{n_p}), C(x_{n,0})) \), reject the proposed merge; set \( (p, n_p) \) as \( (p, n+1) \), and leave \( \tilde{I}(n, r) \) unchanged
  \- Otherwise accept the proposed merge: set \( (p, n_p) \) as \( (p+1, n_p) \) and update \( \tilde{I}(n, r) \) as
    \[
    \tilde{I}(n, p+1) = 1 \quad \text{(add: } T_{n_p,w} u_{p+1}),
    \tilde{I}(n, p) = \tilde{I}(n, 0) = 0 \quad \text{(remove: } T_{n_p,w} u_p, T_{n_a,w})
    \]

**Output:** Return the set of variable-length windows induced by \( \tilde{I}(n, r) \)

The second algorithm we present is based on the dynamic programming approach to adaptive segmentation popular in the audio coding literature [21]–[25]. The basic idea is to fix an additive cost function \( J(\cdot) \), and find an optimal ordered partition function \( \hat{I}(n, r) \) in the sense that it minimizes the sum of individual segment costs \( J(x_{n,r}[t]) \):

\[
\hat{I}^*[n, r] = \arg\min \sum_{n,r} \tilde{I}(n, r) J(x_{n,r}[t]).
\]

Many choices for \( J(\cdot) \) are possible, including rate-distortion cost functions [22], sparsity-inducing measures [14], and the well-known entropy cost of [19], which we employ below.

To formalize our approach, define \( J_n^* \) as the minimum cost among ordered partition functions on \( \{0, 1, \ldots, n \} \), and let \( J_{n,r}^* = J(x_{n,r}[t]) \) represent the cost associated to covering the region \( \{ na, na + 1, \ldots, (n+1)a \} \). The resulting dynamic program requires \( O(N^2) \) iterations and is summarized in Algorithm 2.

Algorithm 2 Adaptation via Dynamic Programming [21]

**Initialization**
- Fix input data \( x \in \mathbb{C}^L \), a Gabor frame \( \mathcal{G}(w, a, b) \) and initialize the cost function \( J_n^* = 0 \)
- For each \( T_{n,w} : n \in \mathbb{Z}_N \) calculate the support set \( \mathcal{G}_n = \{ t : T_{n,w}[t] > T_{n,w}[t], n \neq n' \in \mathbb{Z}_N \} \)

**Dynamic Program**
- For \( n = 0, 1, \ldots, N - 1 \), compute sequentially the \( n \)-th segmental cost and associated boundary by
  \[
  J_n^* = \min_{0 \leq r < n} (J_r^* + J_{n,r}),
  b_n^* = \arg\min_{0 \leq r < n} (J_r^* + J_{n,r}),
  \]
  with \( J_{n,r} \) calculated using signal data supported on \( \mathcal{G}_n \)
- Compute the optimal selection function \( \hat{I}(n, r) \) using \( b_n^* : n \in \mathbb{Z}_N \) via the “backtracking” procedure [21]

**Output:** Return the set of variable-length windows induced by \( \hat{I}(n, r) \)

Note that to preserve cost additivity in the presence of overlapping, non-orthogonal windows, Algorithm 2 evaluates \( J(x_{n,r}[t]) \) on regions smaller than those covered by the corresponding windows, in a manner which recovers the approach of [22] in the block-Fourier case.

Once a set of variable-length windows is obtained via any selection procedure returning an ordered partition function, a local or global modulation structure can be chosen via (15) or (16), respectively, in order to obtain a signal-adaptive superposition frame. In practice, application-specific considerations are likely to play a role in superposition frame selection, and to this end we note that a variety of other algorithms and approaches are possible (see, e.g., [17], [41], [42]).

B. Illustrative Examples

To conclude our investigation of superposition frames, we now consider two illustrative examples that combine Algorithms 1 and 2 with the analysis and reconstruction procedures presented earlier. These examples—a stylized synthetic waveform akin to the example employed in [3] and a phonetically balanced speech utterance from the TIMIT corpus [43]—both exhibit varying time-frequency structure, which in turn motivates signal-adaptive analysis and reconstruction.

Our first example signal \( x \) (Fig. 5 top left) comprises a local and global sinusoidal term, two impulses, and a bump function. We conducted a variety of experiments in which varying levels
of white Gaussian noise $n$ were added to $x$, and Algorithms 1 and 2 were then applied to $y \triangleq x + n$ to obtain signal-adaptive frame analysis coefficients $Y[m,n,r]$ on a global frequency lattice ($M_g = 6000$). Using these as well as fixed-resolution analyses for a range of window lengths, we then applied to $Y[m,n,r]$ both an “oracle” Wiener suppression rule (local signal spectrum estimated by $|X[m,n,r]|^2$) and a two-stage Wiener suppression rule (by appropriately soft-thresholding $|Y[m,n,r]|^2$), and obtained a time-domain reconstruction $\hat{x}$ via the corresponding canonical dual superposition frame.

The remainder of Fig. 5 reports the results of a typical run at 10 dB signal-to-noise ratio (SNR), with the superposition system of Algorithm 1 derived from an initial Gabor system $G(w,a,\cdot)$ comprising a 100-sample Hamming window $w$ and time lattice constant $a = 50$, and that of Algorithm 2 based on a 65-sample Hamming window with $a = 32$. Although we have observed Algorithm 2 to be more noise-robust in practice, and to yield better performance with somewhat shorter initial windows, it may be seen that both algorithms yield broadly similar analyses with respect to dominant signal features at 10 dB SNR. Moreover, over a range of noise levels and fixed-resolution analyses, we have observed improved SNR gains $20 \log_{10} \|y - x\|/\|\hat{x} - x\|$ in both the oracle and two-stage cases, as shown in the bottom-right panels of Fig. 5.

Reconstruction spectrograms $20 \log_{10} |\hat{X}[m,n,r]|$—based on the oracle denoising for visual clarity—are shown in the right-hand panel of Fig. 5. They indicate that, in comparison to an a priori well chosen fixed-resolution analysis using 800-sample Hamming windows with $a = 400$, the onsets and offsets of localized time-frequency features are better preserved by superposition frames. Since the best fixed-resolution window length is not known a priori in practice, the adaptive approach remains attractive, despite the lessening SNR gains obtained in the simple two-stage denoising approach. These results suggest the investigation of more sophisticated denoising schemes, also bearing in mind that in the case of nonstationary noise, the best adaptive analysis may well be SNR-dependent.

We repeated the same battery of tests with our second example signal $x$ (Fig. 6 bottom), a portion of the phonetically-balanced TIMIT speech waveform /train/dr1/fsah0/si1244.wav corresponding to the phrase “...[eye]d and amazed.” This utterance, chosen to illustrate time-varying spectral content typical of speech, contains two plosives ([eye]d, amazed), two steady vowels (and, amazed), and a time-varying diphthong (amazed). The exact phonetic TIMIT segmentation (si1244.phn) is shown between the two spectrogram panels of Fig. 6, which correspond respectively to reconstructions based on an a priori well chosen fixed-resolution (top, 30 ms) and adaptive-resolution (middle, Algorithm 2 starting from 3 ms Hamming windows with $a = 1.5$ ms) oracle-Wiener denoising, respectively, at 10 dB SNR. The bottom panel of Fig. 6 illustrates the corresponding adaptive analysis, which
is seen to agree well with major features of the given TIMIT segmentation; Algorithm 1 also yielded a similar analysis.

Following the same experimental procedure as in the case of Fig. 5, we observed broadly similar results—though with lower overlap. Importantly, however, superposition windows are seen to better preserve vowel onsets and plosives; see in particular the boxed regions of the fixed-resolution spectrogram in the top panel of Fig. 6 corresponding to the two plosives and initial vowel-diphthong onsets in the word “amazed.”

In this manner we see that superposition frames, when coupled with appropriate waveform adaptation criteria, show strong potential for use in a variety of signal-adaptive analysis-synthesis settings. For signal enhancement applications, a natural next step would be to extend the approach of [17], in which adaptive segmentation is used to estimate the local signal spectrum for enhancement purposes, but reconstruction is done using a fixed-resolution time-frequency lattice. A variety of other multi-stage or iterative approaches suggest themselves, given the additional flexibility engendered by the overcomplete, signal-adaptive superposition frames presented in this article.

VIII. DISCUSSION

In this article we have introduced a broad family of adaptive, linear time-frequency representations termed superposition frames, and showed that they admit a host of desirable properties, including fast overlap-add reconstruction akin to standard short-time Fourier techniques. Through a discussion of signal adaptation criteria and multiple examples, the resultant analysis-synthesis systems were seen to provide an effective and practical method for realizing signal-adaptive time-frequency analysis coupled with fast reconstruction.

Relative to other adaptive time-frequency methods, a number of open questions remain. First, while many aspects of our construction admit straightforward extension to other Hilbert spaces of interest such as $L^2(\mathbb{Z})$ or $L^2(\mathbb{R})$, the present article has not addressed window requirements to ensure the existence of upper superposition frame bounds in infinite-dimensional settings, or attempted to characterize the structure of canonical superposition duals in such cases. Second, establishing additional connections to lapped transform constructions, in particular the tight lapped frames recently proposed in [44], seem a promising avenue for further investigation.

APPENDIX

Proof of Lemma 2

To establish (9) for all $w \in C^L$ and $M \geq \text{len}(w)$, expand the left-hand side of (9) as

$$
\sum_{m=0}^{M-1} |\langle x, M_{mb}w \rangle|^2 = \sum_{m=0}^{M-1} \langle x, M_{mb}w \rangle \langle x, M_{mb}w \rangle = \sum_{t=0}^{L-1} \sum_{t'=0}^{L-1} x[t]x[t'] w[t]w[t'] \sum_{m=0}^{M-1} e^{-2\pi im(b(t-t'))/L} = M \sum_{t=0}^{L-1} \sum_{t'=0}^{L-1} \sum_{m=0}^{M-1} \delta_{M,\{t-t'\}} x[t]x[t'] w[t]w[t'] \quad \text{for all } t, t' \geq 0, \text{ whereupon we recover from (33) the right-hand side of (9).}
$$

Now consider all $M \geq \text{len}(w)$ that divide $t - t'$; since $\overline{w[t]w[t']} = 0$ for all $|t - t'| \geq \text{len}(w)$, we need only consider the case $t - t' = 0$, whereupon we recover from (33) the right-hand side of (10).

Proof of Lemma 3

To prove sufficiency, assume that $M_0 = M$, and hence $b_0 = b$. From (9) of Lemma 2 it follows that the right-hand side of (10) may be expanded as

$$
M \sum_{t=0}^{L-1} |x[t]|^2 \left( |w_p[t]|^2 + |w_q[t]|^2 + 2\text{Re}\{\overline{w_p[t]}w_q[t]\} \right),
$$

with the rightmost term nonnegative by Definition 6. Dropping this term from (34) and applying (9) again, this time in the reverse direction, shows that (10) holds for any $M_0 \in \{\text{max}(\text{len}(w_p), \text{len}(w_q)), \ldots, M\}$, thus proving sufficiency.

To prove necessity, assume to the contrary and consider a setting in which $M < \text{len}(w_p + w_q) = L$. Noting that (10) may be stated as $\langle S^{(p,q)}x, x \rangle \leq \langle S^{(p,q)}x, x \rangle$ for positive semi-definite frame operators $S^{(p,q)}$ and $S^{(p+q)}$, assume that elements of the former span $C^L$ and thus form a frame; hence $\langle S^{(p,q)}x, x \rangle > 0$ for all nonzero $x \in C^L$. However, since $M < L$, the $M$ elements of the latter cannot span $C^L$. Hence there exists at least one nonzero $x \in C^L$ such that $\langle S^{(p+q)}x, x \rangle = 0$, thus contradicting the stated inequality.

Proof of Theorem 7

To establish the theorem we directly bound the quantity $\sum_{t \in F(t)} |\langle x, \phi_{m,n,r} \rangle|^2$ from below. To
begin, observe that
\[
\sum_{\phi \in \mathcal{F}(I)} |(x, \phi_{m,n,r})|^2 = \sum_{n=0}^{M_g-1} \sum_{r=0}^{N-1} I[n, r] \left| (x, M_{mbg} T_{na} w_r[t]) \right|^2
\]
with the latter expression above obtained by the expansion of (33). Now, if \( M_g \) divides \( t - t' \), then \( t' = t - k(t) M_g \) for some integer \( k(t) \), whose domain is deduced by observing that \( 0 \leq t - k(t) M_g \leq L - 1 \) and \( 0 \leq t' \leq L - 1 \) together imply that \( k(t) \in \mathcal{K} \). Then, a change of variable for \( t' \) yields the simplification
\[
M_g \sum_{n=0}^{M_g-1} \sum_{r=0}^{N-1} x[t] x[t'] T_{na}(w_r[t] w_r[t']) I[M_g \cdot (t-t')] [t-t'] = 0.
\]

For \( k = 0 \), this quantity can be bounded from below as
\[
M_g \sum_{t=0}^{L-1} |x[t]|^2 \cdot \min_{n \in \mathbb{Z}} \sum_{r=0}^{N-1} I[n, r] \left| T_{na} w_r[t] \right|^2 \geq M_g \|x\|^2 \cdot \min_{n \in \mathbb{Z}} \sum_{r=0}^{N-1} I[n, r] \left| T_{na} w_r[t] \right|^2,
\]
with the remaining terms in \( \mathcal{K} \) handled as follows. Invoking the assumption of a real, nonnegative window \( w \) to simplify the corresponding expression, observe that the terms \( T_{na}(w_r[t] w_r[t-k M_g]) \) are then everywhere nonnegative. The sum of the remaining terms can hence be bounded below by
\[
M_g \sum_{t=0}^{L-1} \sum_{n=0}^{N-1} I[n, r] \left( T_{na} w_r[t] \right)^2 \leq -M_g \|x\|^2 \cdot \min_{n \in \mathbb{Z}} \sum_{r=0}^{N-1} I[n, r] \sum_{t \in \mathbb{Z}} T_{na}(w_r[t] w_r[t-k M_g])
\]
where the second inequality follows by observing that \( \sum_{t=0}^{L-1} x[t+a] x[t] \geq -|a|^2 \) for any \( f \in \mathbb{C}^2 \) and \( s \in \mathbb{Z} \). Thus, we obtain the claimed results since \( \sum_{\phi \in \mathcal{F}(I)} |(x, \phi_{m,n,r})|^2 \) is bounded from below by \( M_g \|x\|^2 \) times
\[
\min_{t \in \mathbb{Z}} \sum_{n=0}^{M_g-1} I[n, r] \left( T_{na} w_r[t] \right)^2 \leq -M_g \|x\|^2 \cdot \min_{n \in \mathbb{Z}} \sum_{r=0}^{N-1} I[n, r] \sum_{t \in \mathbb{Z}} T_{na}(w_r[t] w_r[t-k M_g])
\]

Proof of Theorem 6. To establish the result, first note that \( S_{13} \) is by hypothesis diagonal, and hence by (27), we have
\[
\tilde{\phi}_{m,n,r}[t] = \frac{M_{mbg} T_{na} w_r[t]}{M_g \sum_{n'=0}^{N-1} \sum_{r'=0}^{N-1} I[n', r'] T_{na}(w_{r'}[t])^2}.
\]

If all windows \( \{T_{na} w : n \in \mathbb{Z}\} \) have been merged to yield a single superposition window \( w_{n=1, r} \), whose modules comprise the superposition frame \( \mathcal{F}(I) \) of interest, then the constant overlap-add constraint of (22) applied to (35) immediately implies the result, as both its numerator and denominator yield constants, whose ratio is in turn \( a/(M_g \tilde{w}[0]) \), with \( M_g = \max(L, M) \). Therefore, assume that this is not the case.

To begin, note that (35), together with the neighbor-overlap condition of (28), implies that \( \text{supp}(\phi_{m,n,r}) \subseteq T_{na}(L_R \cup C_R \cup R_c) \). Since these sets are mutually disjoint, we proceed by showing that (35) agrees with
\[
\begin{align*}
\left\{ \phi_{m,n,0}[t] \right\} & \quad \text{if } t \in T_{na} L_R, \\
\left\{ \frac{a}{M_g \tilde{w}[0]} e^{2\pi i mb L t/L} \right\} & \quad \text{if } t \in T_{na} C_R, \\
\left\{ \phi_{m,n+r+1,0}[t] \right\} & \quad \text{if } t \in T_{na} R_c,
\end{align*}
\]
where
\[
\tilde{\phi}_{m,n,0}[t] = \frac{M_{mbg} T_{na} w[t]}{M_g \sum_{n'=0}^{N-1} |w[t - n'a]|^2}.
\]

We now proceed to show that (35) evaluates to (36). First, we have that the numerator of (35) evaluates on \( T_{na} C_R \) to
\[
M_{mbg} T_{na} w_r[t] I_{T_{na} C_R}[t] = M_{mbg} \left( \sum_{n'=0}^{N-1} I_{T_{na} C_R}[t] \right) I_{T_{na} C_R}[t]
\]
and the second equality following from the fact that no windows other than \( T_{na} w_r[t] \) are supported on \( T_{na} C_R \), and the second from (22). Hence we have equality of (35) and (36) on \( T_{na} C_R \).

Applying next the neighbor-overlap condition of (28) and the definition of \( L_R \), we observe that the corresponding numerator term of (35) evaluates to
\[
M_{mbg} T_{na} w_r[t] I_{T_{na} L_R}[t] = M_{mbg} \left( \sum_{n'=0}^{N-1} I_{T_{na} L_R}[t] \right) I_{T_{na} L_R}[t]
\]

Evaluating the denominator of (35) on \( T_{na} L_R \) yields
\[
M_g \sum_{n'=0}^{N-1} \sum_{r'=0}^{N-1} I_{T_{na} L_R}[t] I_{T_{na} L_R}[t] \left| T_{na} w_{r'}[t] \right|^2 = 0,
\]
which may be split into three parts according to index \( n \), including the term \( T_{na} w_r[t] \) as follows:
\[
M_g \left( \sum_{n'=0}^{N-1} \sum_{r'=0}^{N-1} I_{T_{na} L_R}[t] I_{T_{na} L_R}[t] \left| T_{na} w_{r'}[t] \right|^2 \right) \left| T_{na} w_r[t] \right|^2
\]
selects elements from $G^d_{\nu} = \mathcal{G}(w_o, a(r + 1), b_2)$, then $r + 1$ divides $n$ by construction, and (42) follows.

The argument for agreement of the denominators is more delicate, because it is not true that for all $t \in \mathbb{Z}_L$, 
\[
\sum_{n=0}^{N-1} |T_{n,a}w_r[t]|^2 = \sum_{n=0}^{N-1} |T_{n,a}w_r[t]|^2.
\]
(43)
Instead, we show that (43) holds for all $t \in \text{supp}(T_{n,a}w_r)$, which, together with (42), is sufficient to establish (41), and consequently our claimed result.

Let $S_{n,r} \triangleq \text{supp}(T_{n,a}w_r)$, with $\text{supp}(t)$ the corresponding indicator function. Using the same arguments as in the penultimate portion of the proof of Theorem 6 observe that the left-hand side of (43) can be decomposed as follows:
\[
\sum_{n=0}^{N-1} |T_{n,a}w_r[t]|^2 \mathbb{I}_{S_{n,r}}[t]
\]
(43)
Applying the neighbor-overlap requirement of (28) to the right-hand side of (43) then yields
\[
\sum_{n=0}^{N-1} |T_{n,a}w_r[t]|^2 \mathbb{I}_{S_{n,r}}[t]
\]
(43)
thus establishing the equality of (43), and hence the result.

\section*{ACKNOWLEDGMENT}

The authors thank the reviewers for many suggestions that have greatly improved the quality and clarity of this article. In addition, the authors acknowledge helpful discussions co-author with T. F. Quatieri of [1], as well as very generous long-term dialogues with M. Dörfler and L. Rebollo-Neira that have helped this work to achieve its present form.

\section*{REFERENCES}

[1] D. Rudoy, P. Basu, T. F. Quatieri, B. Dunn, and P. J. Wolfe, “Adaptive short-time analysis-synthesis for speech enhancement,” in Proc. IEEE Intl. Conf. Acoust. Speech Signal Process., 2008, pp. 4905–4908. [Online]. Available: http://ssl.eng.umd.edu

[2] D. L. Jones and T. W. Parks, “A high resolution data-adaptive time-frequency representation,” IEEE Trans. Acoust. Speech Signal Process., vol. 38, pp. 2127–2135, 1990.

[3] D. L. Jones and R. G. Baraniuk, “A simple scheme for adapting time-frequency representations,” IEEE Trans. Signal Process., vol. 42, pp. 3530–3535, 1994.

[4] ——, “Adaptive optimal-kernel time-frequency representation,” IEEE Trans. Signal Process., vol. 43, pp. 2361–2371, 1995.

[5] R. N. Czerwinski and D. L. Jones, “Adaptive short-time Fourier analysis,” IEEE Signal Process. Lett., vol. 4, pp. 42–45, 1997.

[6] M. M. Goodwin, Adaptive Signal Models: Theory, Algorithms and Audio Applications. Norwell, MA: Kluwer Academic Publishers, 1998.

[7] S. Mallat, A Wavelet Tour of Signal Processing, 2nd ed. London: Academic Press, 1999.

[8] H. K. Kwon and D. L. Jones, “Improved instantaneous frequency estimation using an adaptive short-time Fourier transform,” IEEE Trans. Signal Process., vol. 48, pp. 2964–2972, 2000.

[9] P. J. Wolfe, S. J. Godsill, and M. Dörfler, “Multi-Gabor dictionaries for audio time-frequency analysis,” in Proc. IEEE Worksh. Appl. Signal Process Audio, 2001, pp. 43–46.

[10] M. Dörfler, “Gabor analysis for a class of signals called music,” Ph.D. dissertation, University of Vienna, July 2002.
11] P. J. Wolfe, S. J. Godsill, and W.-J. Ng, “Bayesian variable selection and regularization for time-frequency surface estimation (with discussion),” *J. R. Statist. Soc. B*, vol. 66, no. 3, pp. 575–589, 2004.

12] F. Jaillet and B. Torresani, “Time-frequency jigsaw puzzle: Adaptive multindow and multilayered Gabor expansions,” *Int. J. Wavelet. Multires. Inf. Process.*, vol. 5, pp. 293–316, 2007.

13] J. Djurovic and L. I. Stankovic, “Adaptive windowed Fourier transform,” *Signal Process.*, vol. 83, pp. 91–100, 2003.

14] A. Nesbit, E. Vincent, and M. D. Plumbley, “Benchmarking flexible adaptive time-frequency transforms for underdetermined audio source separation,” in *Proc. IEEE Intl. Conf. Acoust. Speech Signal Process.*, 2009.

15] P. J. Durka and J. J. Blinowska, “A unified time-frequency parametrization of EEGs,” *IEEE Eng. Med. Biol. Mag.*, vol. 20, pp. 47–53, 2001.

16] E. J. Rothwell, K. M. Chen, and D. P. Nyquist, “An adaptive-window-width short-time Fourier transform for visualization of radar target substructure resonances,” *IEEE Trans. Antenn. Propag.*, vol. 46, pp. 1393–1395, 1998.

17] R. C. Hendriks, R. Heusdens, and J. Jensen, “Adaptive time segmentation for improved speech enhancement,” *IEEE Trans. Audio Speech Lang. Process.*, vol. 14, pp. 2064–2074, 2006.

18] H. S. Malvar, “Lapped transforms for efficient transform/subband coding,” *IEEE Trans. Acoust. Speech Signal Process.*, vol. 38, pp. 969–978, 1990.

19] R. R. Coifman and M. V. Wickerhauser, “Entropy-based algorithms for best basis selection,” *IEEE Trans. Info. Theory*, vol. 38, pp. 713–718, 1992.

20] E. Wesfried and M. V. Wickerhauser, “Adapted local trigonometric transforms and speech processing,” *IEEE Trans. Signal Process.*, vol. 41, pp. 3596–3600, 1993.

21] Z. Xiong, K. Ramchandran, C. Herley, and M. Orchard, “Flexible tree-structured signal expansions using time-varying wavelet packets,” *IEEE Trans. Signal Process.*, vol. 45, pp. 333–245, 1997.

22] P. Prandoni and M. Vetterli, “R/D optimal linear prediction,” *IEEE Trans. Speech Audio Process.*, vol. 8, pp. 646–655, 2000.

23] O. A. Niamut and R. Heusdens, “Optimal time segmentation for overlap-add systems with variable amount of window overlap,” *IEEE Signal Process. Lett.*, vol. 12, pp. 665–668, 2005.

24] R. Heusdens and J. Jensen, “Jointly optimal time segmentation, component selection and quantization for sinusoidal coding of audio and speech,” in *Proc. IEEE Intl. Conf. Acoust. Speech Signal Process.*, vol. 3, 2005, pp. 193–196.

25] C. A. Rødbro, J. Jensen, and R. Heusdens, “Rate-distortion optimal time-segmentation and redundancy selection for VoIP,” *IEEE Trans. Audio Speech Lang. Process.*, vol. 8, pp. 752–763, 2000.

26] Z. Cvetkovic and M. Vetterli, “Oversampled filter banks,” *IEEE Trans. Signal Process.*, vol. 46, pp. 1245–1255, 1998.

27] H. Bölcskei, F. Hlawatsch, and H. G. Feichtinger, “Frame-theoretic analysis of oversampled filter banks,” *IEEE Trans. Signal Process.*, vol. 46, pp. 3256–3268, 1998.

28] J. Kovačević and A. Chebira, “Life beyond bases: The advent of frames (Part II),” *IEEE Signal Process. Mag.*, vol. 24, pp. 115–125, 2007.

29] M. Zibulski and Y. Y. Zeevi, “Discrete multindow Gabor-type transforms,” *IEEE Trans. Signal Process.*, vol. 45, pp. 1428–1442, 1997.

30] S. Li, “Discrete multi-Gabor expansions,” *IEEE Trans. Info. Theory*, vol. 45, pp. 1954–1967, 1999.

31] F. Jaillet, P. Balazs, and M. Dörfler, “Nonstationary Gabor frames,” in *Proc. 8th Intl. Conf. Sampling Theory Appl.*, 2009.

32] S. Qiu and H. G. Feichtinger, “Discrete Gabor structures and optimal representations,” *IEEE Trans. Signal Process.*, vol. 43, pp. 2258–2268, 1995.

33] O. Christensen, *An Introduction to Frames and Riesz Bases*. Boston, MA: Birkhäuser, 2003.

34] C. Heil and D. Walnut, “Continuous and discrete wavelet transforms,” *SIAM Rev.*, vol. 31, pp. 628–666, 1989.

35] S. H. Nawab and T. F. Quatieri, “Short-time Fourier transform,” in *Advanced topics in signal processing*, J. S. Lim and A. Oppenheim, Eds., Prentice Hall, 1988, pp. 289–337.

36] I. Daubechies, A. Grossmann, and Y. Meyer, “Painless nonorthogonal expansions,” *J. Math. Phys.*, vol. 27, pp. 1271–1283, 1986.

37] C. Herley, J. Kovačević, K. Ramchandran, and M. Vetterli, “Tilings of the time-frequency plane: Construction of arbitrary orthogonal bases and fast tiling algorithms,” *IEEE Trans. Signal Process.*, vol. 41, pp. 3341–3359, 1993.

38] G. Wang, “The most general time-varying filter bank and time-varying lapped transforms,” *IEEE Trans. Signal Process.*, vol. 45, pp. 3775–3789, 2006.

39] R. A. Wiggins, “Minimum entropy deconvolution,” *Geoexplorat.*, vol. 16, pp. 21–35, 1978.

40] P. Basu, D. Rudoy, and P. J. Wolfe, “A nonparametric test for stationarity based on local Fourier analysis,” in *Proc. IEEE Intl. Conf. Acoust. Speech Signal Process.*, 2009. [Online]. Available: [http://sisl.seas.harvard.edu](http://sisl.seas.harvard.edu)

41] D. Rudoy, T. F. Quatieri, and P. J. Wolfe, “Time-varying autoregressive tests for multiscale speech analysis,” in *Proc. INTERSPEECH*, 2009, to appear. [Online]. Available: [http://sisl.seas.harvard.edu](http://sisl.seas.harvard.edu)

42] V. Zue, S. Seneff, and J. Glass, “Speech database development at MIT: TIMIT and beyond,” *Speech Commun.*, vol. 9, pp. 351–356, 1990.

43] A. Chebira and J. Kovačević, “Lapped tight frame transforms,” in *Proc. IEEE Intl. Conf. Acoust. Speech Signal Process.*, vol. 3, 2007, pp. 857–860.

44] O. Christensen, *An Introduction to Frames and Riesz Bases*. Boston, MA: Birkhäuser, 2003.