Perfect Reconstruction Two-Channel Wavelet Filter-Banks for Graph Structured Data

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Abstract—In this work we propose the construction of twochannel wavelet filterbanks for analyzing functions defined on the vertices of any arbitrary finite weighted undirected graph. These graph based functions are referred to as graph-signals as we build a framework in which many concepts from the classical signal processing domain, such as Fourier decomposition, signal filtering and downsampling can be extended to graph domain. Especially, we observe a spectral folding phenomenon in bipartite graphs which occurs during downsampling of these graphs and produces aliasing in graph signals. This property of bipartite graphs, allows us to design critically sampled twochannel filterbanks, and we propose quadrature mirror filters (referred to as graph-QMF) for bipartite graph which cancel aliasing and lead to perfect reconstruction. For arbitrary graphs we present a bipartite subgraph decomposition which produces an edge-disjoint collection of bipartite subgraphs. Graph-OMFs are then constructed on each bipartite subgraph leading to "multi-dimensional" separable wavelet filterbanks on graphs. Our proposed filterbanks are critically sampled and we state necessary and sufficient conditions for orthogonality, aliasing cancellation and perfect reconstruction. The filterbanks are realized by Chebychev polynomial approximations.

Note: Code examples from this paper are available at http://biron.usc.edu/wiki/index.php/Graph_Filterbanks

EDICS Category: DSP-WAVL, DSP-BANK, DSP-MULT, DSP-APPL, MLT

I. INTRODUCTION

A. Motivation

Graphs provide a very flexible model for representing data in many domains. Many networks such as biological networks [1], social networks [2], [3] and sensor networks [4], [5] etc. have a natural interpretation in terms of finite graphs with vertices as data-sources and links established based on connectivity, similarity, ties etc. The data on these graphs can be visualized as a finite collection of samples termed as *graph-signals*. For example, graphical models can be used to represent irregularly sampled datasets in Euclidean spaces such as regular grids with missing samples. In many machine learning applications multi-dimensional datasets can be represented as *point-clouds* of vectors and links are established between data sources based on the distance between

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their feature-vectors. In computer vision, meshes are polygon graphs in 2D/3D space and the attributes of the sampled points (coordinates, intensity etc) constitute the graph-signals. The graph-signal formulation can also be used to solve systems of partial differential equations using finite element analysis (grid based solution). The sizes (number of nodes) of the graphs in these applications can be very large, which present computational and technical challenges for the purpose of storage, analysis etc. In some other applications such as wireless sensor-networks, the data-exchanges between far-off nodes can be expensive (bandwidth, latency, energy constraints issues). Therefore, instead of operating on the original graph, it would be desirable to find and operate on smaller graphs with fewer nodes and data representing a smooth approximation of the original data. Moreover, such systems need to employ localized operations which could be computed at each node by using data from a small neighborhood of nodes around it. Multi-channel wavelet filterbanks, widely used as a signal processing tool for the sparse representation of signals, possess both these features (i.e. smooth approximations and localized operations). For example, a two channel wavelet transform splits the sample space into an approximation subspace which contains a smoother (coarser) version of the original signal and a detail subspace containing additional details required to perfectly reconstruct the original signal. A discussion of the construction and analysis of wavelet filterbanks for regular signals can be found in standard textbooks such as [6]. While wavelet transform-based techniques would seem well suited to provide efficient local analysis, a major obstacle to their application to graphs is that these, unlike images, are not regularly structured. For graphs traditional notions of dimensions along which to filter the data do not hold.

Researchers have recently focused on developing localized transforms specifically for data defined on graphs. Crovella and Kolaczyk [2] designed wavelet like functions on graphs which are localized in space and time. These graph functions $\psi_{j,k}$ are composed of either shifts or dilations of a single generating function ψ . Wang and Ramchandran [5] proposed graph dependent basis functions for sensor network graphs, which implement an invertible 2-channel like filter-bank. There exists a natural spectral interpretation of graph-signals in terms of eigen-functions and eigen-values of graph Laplacian matrix L. Maggioni and Coifman [7] introduced "diffusion wavelets" as the localized basis functions of the eigenspaces of the dyadic powers of a diffusion operator. Hammond et. al. [8] construct a class of wavelet operators in the graph

¹more generally, it could be any sparse approximation of the original data.

spectral domain, i.e., the space of eigenfunctions of the graph Laplacian matrix L. These eigenfunctions provide a spectral decomposition for data on a graph similar to the Fourier transform for standard signals. A common drawback of all of these filterbank designs is that they are not critically sampled : the output of the transform is not downsampled and there is oversampling by a factor equal to the number of channels in the filterbank. Unlike classical wavelet transforms which have well-understood downsampling/upsampling operations, there is no obvious way in graphs to downsample nodes in a regular manner, since the neighboring nodes vary in number. Lifting based wavelet transforms have been proposed in [9], [10] for graphs in Euclidean Space and in our previous work for trees in [4], [11] and for general graphs in [12]. These transforms are critically sampled and invertible by construction. However the design requires splitting the vertex set of the graph into two disjoint sets and the transform is computed only on the links between nodes in different sets. Thus links between nodes in same set are not utilized by the transform.

Our contribution in this paper is to introduce a theory behind sampling operations on graphs, which leads us to the design of critically-sampled wavelet-filterbanks on graphs. We describe a downsample then upsample (DU) operation on graphs in which a set of nodes in the graph are first downsampled (removed) and then upsampled (replaced) by inserting zeros. This work stems from our recent results in [13], where we showed that downsampling for graph-signals defined on k-regular bipartite graphs is governed by a Nyquist-like theorem. In this paper, we extend the results presented in [13] to all undirected bipartite graphs and show that in these graphs, the DU operations lead to a spectral decomposition of the graph-signal where spectral coefficients are reproduced at mirror graph-frequencies around a central frequency. This is a phenomenon we term as spectrum folding in graphs as it is analogous to the frequency-folding or "aliasing" effect for regular one-dimensional signals. We utilize this property to propose two-channel filterbanks on bipartite graphs which are critically sampled and provide necessary and sufficient conditions for aliasing cancellation, perfect-reconstruction and orthogonality in these filterbanks. As a practical solution we propose a graph-quadrature mirror filterbank (referred to as graph-QMF) design for bipartite graphs which has all the above mentioned properties. However, the exact realizations of the graph-QMF filters do not have well-localized support on the graph and therefore we implement polynomial approximations of these filters which are locally supported around each node (at the cost of small reconstruction error and loss of orthogonality). For arbitrary graphs, we formulate a bipartite subgraph decomposition problem well known to the graphtheory community. The decomposition provides us an edgedisjoint collection of K bipartite subgraphs, each with the same vertex set \mathcal{V} and whose union is the original graph. Each of these subgraphs is then used as a separate "dimension" to filter and downsample leading to a K-dimensional separable wavelet filterbank design. To the best of our knowledge no such invertible and critically sampled two-channel filter-bank designs have been proposed for arbitrary graphs before. The outline for the rest of the paper is as follows: we describe

the basic framework to understand graph-based transforms in Section II. In this section we also describe and evaluate some of the existing work on wavelet-like transforms on graph. In Section III we propose our solution and in Section IV we demonstrate the utility of proposed filterbanks by conducting some experiments. Finally, in Section V, we conclude and describe our future work.

II. PRELIMINARIES

We use the common convention of representing matrices and vectors with bold letters, sets with calligraphic capital letters and scalars with normal letters. A graph can be denoted as $G = (\mathcal{V}, E)$ with vertices (or nodes) in set \mathcal{V} and links (or edges) as tuples (i, j) in E. We only consider undirected graphs without self-loops in our work. The size of the graph $N = |\mathcal{V}|$ is the number of nodes and geodesic distance metric is given as d(v, m). The j-hop neighborhood $\mathcal{N}_{i,n} = \{v \in \mathcal{V} : v \in$ $d(v, n) \leq j$ of node n is the set of all nodes which are at most j-hop distance away from node n. Algebraically, a graph can be represented with the node-node adjacency matrix A such that the element A(i,j) is the weight of the edge between node i and j (0 if no edge). The value d_i is the degree of node i, which is the sum of weights of all edges connected to node i, and $\mathbf{D} = diag(\{d_i\})$ denotes the diagonal degree matrix whose i^{th} diagonal entry is d_i . The Laplacian matrix of the graph is defined as L = D - A and has a normalized form $\mathcal{L} = \mathbf{I} - \mathbf{D}^{-1/2} \mathbf{A} \mathbf{D}^{-1/2}$, where \mathbf{I} is the identity matrix. We denote $\langle \mathbf{f}_1, \mathbf{f}_2 \rangle$ as the inner-product between vectors \mathbf{f}_1 and \mathbf{f}_2 .

A. Spatial Representation of Graph Signals

A graph signal is a real-valued scalar function $f: \mathcal{V} \to \mathbb{R}$ defined on graph $G = (\mathcal{V}, E)$ such that f(v) is the sample value of function at vertex $v \in \mathcal{V}$. On a finite graph, the graph-signal can be viewed as a sequence or a vector $\mathbf{f} = [f(0), f(1), ..., f(N)]^t$, where the order of arrangement of the samples in the vector is arbitrary and neighborhood (or nearness) information is provided separately by the adjacency matrix \mathbf{A} . Graph-signals can, for example, be a set of measured values by sensor network nodes [5] or traffic measurement samples on the edges of an Internet graph [2] or information about the actors in a social network. Further, a graph based transform is defined as a linear transform $\mathbf{T}: \mathbb{R}^N \to \mathbb{R}^M$ applied to the N-node graph-signal space, such that the operation at each node n is a linear combination of the value of the graph-signal f(n) at the node n and the values f(m) on nearby nodes $m \in \mathcal{N}_{j,n}$, i.e.,

$$y(n) = T(n,n)f(n) + \sum_{m \in \mathcal{N}_{j,n}} T(n,m)f(m)$$
 (1)

In analogy to the 1-D regular case, we would sometimes refer to graph-transforms as graph-filters and the elements T(n,m) for m=1,2,...N as the filter coefficients at the n^{th} node. A graph transform is said to be strictly j-hop localized in the

²The extension to complex or vector sample values f(v) is possible but is not considered in this work.

spatial domain of the graph if the filter coefficients T(n,m) are zero beyond the j-hop neighborhood of each node n. Note that spatial localization can also be applied in a weaker sense in which filter coefficients T(n,m) decay sharply in magnitude beyond j-hop neighborhood of node n.

B. Spectral Representation of Graph Signals

The Laplacians \mathbf{L} and \mathcal{L} are both symmetric positive semidefinite matrices and therefore, from the spectral projection theorem, there exists a real unitary matrix \mathbf{U} which diagonalizes \mathcal{L} , such that $\mathbf{U}^t \mathcal{L} \mathbf{U} = \mathbf{\Lambda} = \mathbf{diag}\{\lambda_i\}$ is a non-negative diagonal matrix. This leads to an *eigenvalue decomposition* of matrix \mathcal{L} given as

$$\mathcal{L} = \mathbf{U}\boldsymbol{\Lambda}\mathbf{U}^{t} = \sum_{i=1}^{N} \lambda_{i}\mathbf{u}_{i}\mathbf{u}_{i}^{t}, \tag{2}$$

where the eigenvectors $\mathbf{u}_1, \mathbf{u}_2, ..., \mathbf{u}_N$, which are columns of \mathbf{U} form a basis in \mathbb{R}^N and the corresponding eigenvalues $\sigma(G) = \{0 \leq \lambda_1 \leq \lambda_2 \dots \leq \lambda_N\}$ represent N orthogonal eigen-spaces V_{λ_i} with projection matrices $\mathbf{u}_i \mathbf{u}_i^t$. Thus, every graph-signal $\mathbf{f} \in \mathbb{R}^N$ can be decomposed into a linear combination of eigenvectors \mathbf{u}_i given as $\mathbf{f} = \sum_{n=1}^N \bar{f}(n)\mathbf{u}_n$. It has been shown in [14], [15] that the eigenvectors of Laplacian matrix provide a harmonic analysis of graph signals which gives a Fourier-like interpretation. The eigenvectors act as the natural vibration modes of the graph, and the corresponding eigenvalues as the associated graph-frequencies. The spectrum $\sigma(G)$ of a graph is defined as the set of eigen-values of its normalized Laplacian matrix and it is always a subset of closed set [0, 2] for any graph G. Any eigenvector \mathbf{u}_{λ} is considered to be a low pass eigenvector if the magnitude of the corresponding eigenvalue λ is small, i.e., close to 0. Similarly, an eigenvector is a high-pass eigenvector if its eigenvalue is large, i.e., close to the highest graph-frequency. ¹ The graph Fourier transform (GFT), denoted as $\bar{\mathbf{f}}$, is defined in [8] as the projections of a signal f on the graph G onto the eigenvectors of G, i.e.,

$$\bar{f}(\lambda) = \langle \mathbf{u}_{\lambda}, \mathbf{f} \rangle = \sum_{i=1}^{N} f(i)u_{\lambda}(i).$$
 (3)

Note that GFT is an energy preserving transform and a signal can be considered low-pass (or high-pass) if the energy $|\bar{f}(\lambda)|^2$ of the GFT coefficients is mostly concentrated on the low-pass (or high-pass) eigenvectors. In case of eigenvalues with multiplicity greater than 1 (say $\lambda_1 = \lambda_2 = \lambda$) the eigenvectors $\mathbf{u}_1, \mathbf{u}_2$ are unique up to a unitary transformation in the eigenspace $V_{\lambda} = V_{\lambda_1} = V_{\lambda_2}$. In this case we can choose $\lambda_1 \mathbf{u}_1 \mathbf{u}_1^t + \lambda_2 \mathbf{u}_2 \mathbf{u}_2^t = \lambda \mathbf{P}_{\lambda}$ where \mathbf{P}_{λ} is the projection matrix for eigenspace V_{λ} . Note that for all symmetric matrices, the

 1 The mapping $\mathbf{u}_n \to \mathcal{V}$ associates the real numbers $u_n(i), i = \{1, 2, ..., N\}$, with the vertices \mathcal{V} of G. The numbers $u_n(i)$ will be positive, negative or zero. The frequency interpretation of eigenvectors can thus be understood in terms of number of zero-crossings (pair of nodes with different signs) of eigenvector \mathbf{u}_n on the graph G. For any finite graph the eigenvectors with large eigenvalues have more zero-crossings (hence high-frequency) than eigenvectors with small eigenvalues. These results are related to 'nodal domain theorems' and readers are directed to [16] for more details.

dimension of eigenspace V_{λ} (geometric multiplicity) is equal to the multiplicity of eigenvalue λ (algebraic multiplicity) and the spectral decomposition in (2) can be written as

$$\mathcal{L} = \sum_{\lambda \in \sigma(\mathbf{G})} \lambda \sum_{\lambda_{i} = \lambda} \mathbf{u}_{i} \mathbf{u}_{i}^{t} = \sum_{\lambda \in \sigma(\mathbf{G})} \lambda \mathbf{P}_{\lambda}.$$
 (4)

The eigenspace projection matrices are idempotent and P_{λ} and P_{γ} are orthogonal if λ and γ are distinct eigenvalues of the Laplacian matrix, i.e.,

$$\mathbf{P}_{\lambda}\mathbf{P}_{\gamma} = \delta(\lambda - \gamma)\mathbf{P}_{\lambda},\tag{5}$$

where $\delta(\lambda)$ is the Kronecker delta function.

C. Downsampling in Graphs

We define the downsampling operation β_H on the graph $G=(\mathcal{V},E)$ as choosing a subset $H\subset\mathcal{V}$ such that all samples of the graph signal \mathbf{f} , corresponding to indices not in H, are discarded. A subsequent upsampling operation projects the downsampled signal back to original \mathbb{R}^N space by inserting zeros in place of discarded samples in $H^c=L$. Given such a set H we define a downsampling function $\beta_H\in\{-1,+1\}$ given as

$$\beta_H(n) = \begin{cases} 1 & \text{if } n \in H \\ -1 & \text{if } n \notin H \end{cases}$$
 (6)

and a diagonal downsampling matrix $\mathbf{J}_{\beta_H} = diag\{\beta_H(n)\}$. The overall 'downsample then upsample' (DU) operation can then be algebraically represented as

$$f_{du}(n) = \frac{1}{2}(1 + \beta_H(n))f(n)$$
 (7)

and in matrix form as

$$\mathbf{f}_{du} = \frac{1}{2} (\mathbf{I} + \mathbf{J}_{\beta_H}) \mathbf{f} \tag{8}$$

Note that \mathbf{J}_{β_H} is a symmetric matrix such that $\mathbf{J}_{\beta_H}^2 = \mathbf{I}$ (identity matrix). Since the graph-signal after DU operation also belongs to \mathbb{R}^N , it too has a GFT decomposition $\bar{\mathbf{f}}_{du}$ according to (3). The relationship between the GFTs of \mathbf{f} and \mathbf{f}_{du} is given as:

$$\bar{f}_{du}(l) = \langle \mathbf{u}_l , \mathbf{f}_{du} \rangle = \frac{1}{2} (\langle \mathbf{u}_l , \mathbf{f} \rangle + \langle \mathbf{u}_l , \mathbf{J}_{\beta_H} \mathbf{f} \rangle)$$
 (9)

The inner-product $< \mathbf{u}_l$, $\mathbf{J}_{\beta_H} \mathbf{f} >$ can also be written as $< \mathbf{J}_{\beta_H} \mathbf{u}_l$, $\mathbf{f} >$, which represents the projection of input signal \mathbf{f} onto a *deformed* eigenvector $\mathbf{J}_{\beta_H} \mathbf{u}_l$. We define this projection as a *deformed* spectral coefficient $\mathbf{f}^d(l)$ and (9) can be written as:

$$\bar{f}_{du}(l) = \frac{1}{2}(\bar{f}(l) + \langle \mathbf{J}_{\beta_H} \mathbf{u}_l , \mathbf{f} \rangle) = \frac{1}{2}(\bar{f}(l) + \bar{f}^d(l))$$
 (10)

In case of bipartite graphs, the spectrum of the graph is symmetric and the deformed eigenvectors are also the eigenvectors of the same graph. This phenomenon, termed as *spectral folding*, forms the basis of our two-channel filterbank framework, and will be described in detail in Section III.

D. Two-Channel Filterbanks on Graph

A two-channel wavelet filterbank on a graph provides a decomposition of any graph-signal into a lowpass (smooth) graph-signal and a highpass (detail) graph-signal component. The two channels of the filterbanks are characterized by the graph-filters $\{\mathbf{H}_i, \mathbf{G}_i\}_{i \in \{0,1\}}$ and the downsampling operations β_H and β_L as shown in Figure 1. The transform \mathbf{H}_0 acts as a lowpass filter, i.e., it transfers the contributions of the low-pass graph-frequencies, which are below some cut-off and attenuates significantly the graph-frequencies which are above the cut-off. The highpass transform \mathbf{H}_1 does the opposite of a low-pass transform, i.e, it attenuates significantly, the graph-frequencies below some cut-off frequency. The filtering operations in each channel are followed by downsampling operations β_H and β_L , which means that the nodes with membership in the set H store the output of highpass channel while the nodes in the set L store the output of lowpass channel. For critically sampled output we have: |H| + |L| = N. Using (8),

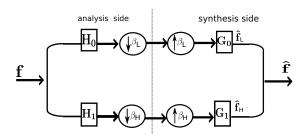


Fig. 1: Block diagram of a two-channel wavelet filterbank on graph.

it is easy to see from Figure 1 that the output signals in the lowpass and highpass channels, after reconstruction are given as

$$\hat{\mathbf{f}}_{L} = \frac{1}{2}\mathbf{G}_{0}(\mathbf{I} + \mathbf{J}_{\beta_{L}})\mathbf{H}_{0}\mathbf{f}$$

$$\hat{\mathbf{f}}_{H} = \frac{1}{2}\mathbf{G}_{1}(\mathbf{I} + \mathbf{J}_{\beta_{H}})\mathbf{H}_{1}\mathbf{f},$$
(11)

respectively. The overall output $\hat{\mathbf{f}}$ of the filterbank is the sum of outputs of the two channels, i.e., $\hat{\mathbf{f}} = \hat{\mathbf{f}}_L + \hat{\mathbf{f}}_H = \mathbf{T}\mathbf{f}$, where \mathbf{T} is the overall transfer function of the filterbank given as:

$$\mathbf{T} = \frac{1}{2}\mathbf{G}_{0}(\mathbf{I} + \mathbf{J}_{\beta_{L}})\mathbf{H}_{0} + \frac{1}{2}\mathbf{G}_{1}(\mathbf{I} + \mathbf{J}_{\beta_{H}})\mathbf{H}_{1}$$
(12)
$$= \underbrace{\frac{1}{2}(\mathbf{G}_{0}\mathbf{H}_{0} + \mathbf{G}_{1}\mathbf{H}_{1})}_{\mathbf{T}_{eq}} + \underbrace{\frac{1}{2}(\mathbf{G}_{0}\mathbf{J}_{\beta_{L}}\mathbf{H}_{0} + \mathbf{G}_{1}\mathbf{J}_{\beta_{H}}\mathbf{H}_{1})}_{\mathbf{T}_{alias}}$$

where \mathbf{T}_{eq} is the transfer function of the filterbank without the DU operation and \mathbf{T}_{alias} is another transform which arises primarily due to the downsampling in the two channels. For perfect reconstruction \mathbf{T} should be equal to identity which can be ensured by requiring \mathbf{T}_{eq} to be a scalar multiple of identity and $\mathbf{T}_{alias} = \mathbf{0}$. Thus the two-channel filterbank on a graph provides distortion-free perfect reconstruction if

$$G_0 J_{\beta_L} H_0 + G_1 J_{\beta_H} H_1 = 0$$

 $G_0 H_0 + G_1 H_1 = cI$ (13)

In order to design perfect reconstruction filterbanks we need

 2 Note that in the regular signal domain the two most common patterns of critically sampled output are i) $H=L=\{0,2,4,\ldots\}$, where even set of nodes store the output of both channels and ii) $L=\{0,2,4,\ldots\}$ and $H=\{1,3,5,\ldots\}$, where each node stores the output of only one of the channel.

to determine a) how to design filtering operations \mathbf{H}_i , \mathbf{G}_i , $i = \{0,1\}$, and b) the downsampling functions β_L and β_H . In Section III, we show that the spectral folding phenomenon in bipartite graphs leads to an aliasing interpretation of (13) and we design filterbanks which cancel aliasing and lead to perfect reconstruction of any graph-signal. Before explaining our approach, we briefly analyze and evaluate some of the existing graph based transforms, by representing them using the framework we just introduced.

E. Existing Designs

Existing designs of wavelet-like filterbanks on the graph can be divided into two types, namely, spatial and spectral designs. In order to understand these designs we introduce some additional notation. We define $\partial \mathcal{N}_{h,k}$ to be an h-hop neighborhood ring around node k (i.e., the set of all nodes which are exactly h hops away from node k), a j-hop adjacency matrix \mathbf{A}_j s.t. $\mathbf{A}_j(n,m)=1$ only if $m\in\mathcal{N}_{j,n}$, a j-hop diagonal degree matrix with $D_j(k,k)=|\mathcal{N}_{j,k}|$ s.t. $d_{j,k}=|\mathcal{N}_{j,k}|$ and a j-hop uniform Laplacian matrix $\mathbf{L}_j=\mathbf{D}_j-\mathbf{A}_j$. Similarly we define a ring adjacency matrix $\partial \mathbf{A}_h$ such that $\partial \mathbf{A}_j(n,m)=1$ only if $m\in\partial\mathcal{N}_{j,n}$ and corresponding ring degree matrix $\partial \mathbf{D}_h=diag\{\partial d_{j,k}\}$ s.t. $\partial d_{j,k}=|\partial \mathcal{N}_{h,k}|$.

1) Spatial Designs: Wang and Ramchandran [5] proposed spatially localized graph transforms for sensor network graphs with binary links, (i.e. links which have weight either 0 or 1). The transforms proposed in [5] either compute a weighted average given as

$$y(n) = (1 - a + \frac{a}{d_{j,k} + 1})x(n) + \sum_{m \in \mathcal{N}_{j,n}} \frac{a}{d_{j,k} + 1}x(m), (14)$$

or a weighted difference given as

$$y(n) = (1 + b - \frac{b}{d_{j,k} + 1})x(n) - \sum_{m \in \mathcal{N}_{i,n}} \frac{b}{d_{j,k} + 1}x(m), (15)$$

in a j-hop neighborhood around each node in the graph. The corresponding transform matrices can be represented for a given j as

$$\mathbf{T}_{j} = \mathbf{I} - a(\mathbf{I} + \mathbf{D}_{j})^{-1} \mathbf{L}_{j}$$

$$\mathbf{S}_{j} = \mathbf{I} + b(\mathbf{I} + \mathbf{D}_{j})^{-1} \mathbf{L}_{j}.$$
 (16)

This approach intuitively defines a two-channel wavelet filterbank on the graph consisting of two types of linear filters: a) *approximation filters* as given in (14) and b) *detail filters* as given in (15). However, these transforms are oversampled and produce output of the size twice that of the input. Further none of the transforms can be called a wavelet filter since both transforms have a non-zero DC response.

Crovella and Kolaczyk [2] designed wavelet like transforms on graphs which are localized in space. They defined a collection of functions $\psi_{j,n}: \mathcal{V} \to \mathbb{R}$, localized with respect to a range of scale/location indices (j,n), which at a minimum satisfy $\sum_{m \in \mathcal{V}} \psi_{j,n}(m) = 0$ (i.e. a zero DC response). Each function $\psi_{j,n}$ is constant within hop rings $\partial \mathcal{N}_{h,n}$ and can be written as:

$$y(n) = a_{j,0}x(n) + \sum_{h=1}^{j} \sum_{m \in \mathcal{N}_{i}} \frac{a_{j,h}}{\partial d_{j,n}} x(m)$$
 (17)

In matrix form the j-hop wavelet transform \mathbf{T}_j can be written as:

$$\mathbf{T}_{j} = a_{j,0}\mathbf{I} + a_{j,1}\partial\mathbf{D}_{1}^{-1}\partial\mathbf{A}_{1} + ...a_{j,j}\partial\mathbf{D}_{j}^{-1}\partial\mathbf{A}_{j}$$
 (18)

Further, the constants $a_{j,h}$ satisfy $\sum_{h=0}^{h=j} a_{j,h} = 0$, which allows the wavelet filters to have zero DC response. Though these transforms are local and provide a multi-scale summarized view of the graph, they do not have approximation filters and are not invertible in general.

Lifting based wavelet transforms for graphs have been proposed in [9], [4], [12], [10] and provide a natural way of constructing local two-channel critically sampled filterbanks on graph-signals. In this approach the vertex set is first partitioned into sets of even and odd nodes $\mathcal{V} = \mathcal{O} \cup \mathcal{E}$. The odd nodes compute their *prediction* coefficients using their own data and data from their even neighbors followed by even nodes computing their *update* coefficients using their own data and prediction coefficient of their neighboring odd nodes. The equivalent transform in matrix-form can be written as:

$$\mathbf{T}^{lift} = \overbrace{\begin{bmatrix} \mathbf{I}_{\mathcal{O}} & 0 \\ \mathbf{U} & \mathbf{D}_{\mathcal{E}} \end{bmatrix}}^{update} \overbrace{\begin{bmatrix} \mathbf{D}_{\mathcal{O}} & -\mathbf{P} \\ 0 & \mathbf{I}_{\mathcal{E}} \end{bmatrix}}^{predict}$$
(19)

where $\mathbf{D}_{\mathcal{O}}$ and $\mathbf{D}_{\mathcal{E}}$ are diagonal matrices of size $|\mathcal{O}|$ and $|\mathcal{E}|$ respectively. Although the lifting scheme can be applied to any arbitrary graph, the design is equivalent to simplification of the graph to a bipartite (2-colorable) graph, given that nodes of the same color/parity cannot use each other's data even if they are connected by an edge. This results in edge losses.

2) Spectral Designs: Maggioni and Coifman [7] introduced "diffusion wavelets", a general theory for wavelet decompositions based on compressed representations of powers of a diffusion operator (such as Laplacian). Their construction interacts with the underlying graph or manifold space through repeated applications of a diffusion operator T, such as the graph Laplacian L. The localized basis functions at each resolution level are orthogonalized and downsampled appropriately to transform sets of orthonormal basis functions through a variation of the Gram-Schmidt orthonormalization (GSM) scheme. Although this local GSM method orthogonalizes the basis functions (filters) into well localized 'bump-functions' in the spatial domain, it does not provide guarantees on the size of the support of the filters it constructs. Further the diffusion wavelets form an over-complete basis and there is no simple way of representing the corresponding transform T.

Hammond et al [8] defined spectral graph wavelet transforms that are determined by the choice of a kernel function $g: \mathbb{R}^+ \to \mathbb{R}^+$. The kernel $g(\lambda)$ is a continuous bandpass function in spectral domain with g(0) = 0 and $\lim_{\lambda \to \infty} g(\lambda) = 0$. The corresponding wavelet operator $\mathbf{T}_g = g(\mathcal{L}) = \mathbf{Ug}(\Lambda)\mathbf{U}^t$ acts on a graph signal \mathbf{f} by modulating each Fourier mode as

$$\mathbf{T}_g \mathbf{f} = \sum_{k=1}^{N} g(\lambda_k) \bar{f}(k) \mathbf{u}_k$$
 (20)

The kernel can be scaled as $g(t\lambda)$ by a continuous scalar t. For spatial localization, the authors design filters by approximating the kernels $g(\lambda)$ with smooth polynomials functions.

The approximate transform with polynomial kernel of degree k is given by $\mathbf{T}_{\hat{g}} = \hat{g}(L) = \sum_{l=0}^k a_l L^l$ and is exactly khop localized in space. By construction the spectral wavelet transforms have zero DC response, hence in order to stably represent the low frequency content of signal f a second class of kernel function $h: \mathbb{R}^+ \to \mathbb{R}^+$ is introduced which acts as a lowpass filter, and satisfies h(0) > 0 and $\lim_{\lambda \to \infty} h(\lambda) = 0$. Thus a multi-channel wavelet transform can be constructed from the choice of a low pass kernel $h(\lambda)$ and J bandpass kernels $\{g(t_1\lambda),...,g(t_J\lambda)\}$ and it is been shown that the perfect reconstruction of the original signal is assured if the quantity $G(\lambda) = h(\lambda)^2 + \sum_{k=1}^{J} g(t_k \lambda)^2 > 0$ on the spectrum of \mathcal{L} (i.e., at the N eigenvalues of \mathcal{L}). However, these transforms are overcomplete, for example, a J-scale decomposition of graph-signal of size N produces (J+1)Ntransform coefficients. As a result, the transform is invertible only by the least square projection of the output signal onto a lower dimension subspace.

To conclude this section, Table I presents a summary of existing methods and their properties.

Method	DC response	CS	PR	OE	GS
Wang & Ramchan-	non-zero	No	Yes	No	No
dran [5]					
Crovella &	zero	No	No	No	No
Kolaczyk [2]					
Lifting Scheme [12]	zero for wavelet basis	Yes	Yes	No	Yes
Diffusion Wavelets [7]	zero for wavelet basis	No	Yes	Yes	No
Spectral Wavelets [8]	zero for wavelet basis	No	Yes	No	No
Proposed graph-QMF	zero (when degree-	Yes	Yes	Yes	No
filterbanks	normalized) for				
	wavelet basis				

TABLE I: Evaluation of graph wavelet transforms. CS: Critical Sampling, PR: Perfect Reconstruction, OE: Orthogonal Expansion, GS: Requires Graph Simplification.

III. PROPOSED SOLUTION

Our proposed two-channel critically sampled graph wavelet filterbanks are shown in Figure 1. For DU operations we choose a specific downsampling pattern in which sets H and L provide a bipartition of the graph nodes i.e., $H \cap L = \emptyset$ and $H \cup L = \mathcal{V}$). This implies that downsampling functions $\beta_L(n) = -\beta_H(n) = \beta(n)$ and the nodes in L store the output of the lowpass channel whereas the nodes in the complement set H store the output of the highpass channel. The overall output after filtering and downsampling operations in both channels is critically sampled. For designing wavelet filters on graphs we exploit similar concepts of spectral decomposition as in [8]. Because of this, it is useful to define analysis wavelet filters \mathbf{H}_0 and \mathbf{H}_1 in terms of spectral kernels $h_0(\lambda)$ and $h_1(\lambda)$ respectively. Thus given the eigen-space decomposition of Laplacian matrix \mathcal{L} as in (4), the analysis filters can be represented as:

$$\mathbf{H}_{0} = h_{0}(\mathcal{L}) = \sum_{\lambda \in \sigma(G)} h_{0}(\lambda) \mathbf{P}_{\lambda}$$

$$\mathbf{H}_{1} = h_{1}(\mathcal{L}) = \sum_{\lambda \in \sigma(G)} h_{1}(\lambda) \mathbf{P}_{\lambda}$$
(21)

Since the Laplacian matrix \mathcal{L} is real and symmetric, the filters designed in (21) are also real and symmetric. As described

in Section II-D, the spectral decomposition of the output of a DU operation with downsampling function β yields a set of original signal coefficients and a set of deformed signal coefficients which are generated by the projection of the original signal onto deformed eigenvectors. In what follows, we take the special case of bipartite graphs for which the DU operation on any eigenvector produces an alias eigenvector at a mirror eigenvalue, a phenomenon, which is analogous to the "aliasing" effect observed in DU operations in regular signal domain. This property of bipartite graphs, allows us to express the perfect reconstruction conditions for the two channel graph-filterbanks, as given in (13), in simple terms. Subsequently, we state necessary and sufficient conditions for a two-channel graph filter-bank, designed using spectral transforms, to provide aliasing-cancellation, perfect reconstruction and an orthogonal decomposition of any graphsignal and propose a solution similar to quadrature mirror filters (OMF) in regular signal domain which satisfies all of the above conditions. For arbitrary graphs, we formulate a bipartite subgraph decomposition problem that provides us with an edge-disjoint collection of bipartite subgraphs whose union is G. A wavelet filterbank can be constructed on each of these subgraphs leading to a multi-dimensional separable wavelet filterbank on any arbitrary graph. Finally, we propose a multi-resolution implementation in which the proposed filterbanks can be recursively applied to the downsampled output coefficients of each channel.

A. Downsampling in bipartite graphs

A bipartite graph G=(L,H,E) is a graph whose vertices can be divided into two disjoint sets L and H, such that every link connects a vertex in L to one in H. Bipartite graphs are also known as two-colorable graphs since the vertices can be colored perfectly into two colors so that no two connected vertices are of the same color. Examples of bipartite graphs include tree graphs, cycle graphs and planar graphs with even degrees. In our analysis, we use the normalized form of the Laplacian matrix $\mathcal{L} = \mathbf{D}^{-1/2}\mathbf{L}\mathbf{D}^{-1/2}$ for the bipartite graph, which in the case of regular graphs has the same set of eigenvectors as \mathbf{L} . The normalization reweighs the edges of graph G so that the degree of each node is equal to 1. To understand the spectral interpretation of DU operations in bipartite graphs, the following properties of bipartite graphs are useful:

Lemma 1 ([17, Lemma 1.8]): The following statements are equivalent for any graph G:

- 1) G is bipartite with bipartitions H and L.
- 2) The spectrum of $\mathcal{L}(\mathbf{G})$ is symmetric about 1 and the minimum and maximum eigenvalues of $\mathcal{L}(\mathbf{G})$ are 0 and 2 respectively.
- 3) If $\mathbf{u} = \begin{bmatrix} \mathbf{u}_1^T & \mathbf{u}_2^T \end{bmatrix}^T$ is an eigenvector of \mathcal{L} with eigenvalue λ with \mathbf{u}_1 indexed on H and \mathbf{u}_2 indexed on L (or vice-versa) then the deformed eigenvector $\hat{\mathbf{u}} = \begin{bmatrix} \mathbf{u}_1^T & -\mathbf{u}_2^T \end{bmatrix}^T$ is also an eigenvector of \mathcal{L} with eigenvalue 2λ .

Based on these properties of bipartite graphs we state our key result below:

Proposition 1: Given a bipartite graph G = (L, H, E) with Laplacian matrix \mathcal{L} , if we choose downsampling function β as β_H or β_L as defined in (6), and if \mathbf{P}_{λ} is the projection matrix corresponding to the eigenspace V_{λ} , then

$$\mathbf{J}_{\beta}\mathbf{P}_{\lambda} = \mathbf{P}_{2-\lambda}\mathbf{J}_{\beta}.\tag{22}$$

Alternatively, if \mathbf{u}_{λ} is an eigen-vector of \mathcal{L} with eigenvalue λ then $\mathbf{J}_{\beta}\mathbf{u}_{\lambda}$ is also an eigen-vector of \mathcal{L} with eigen-value $2-\lambda$.

Proof: Let λ be an eigenvalue of G with multiplicity k. This implies that there exists an orthogonal set of k eigenvectors $\{\mathbf{u}_i\}_{\lambda_i=\lambda}$ of Laplacian matrix \mathcal{L} with eigenvalue λ . The projection matrix \mathbf{P}_{λ} corresponding to λ is given by $\mathbf{P}_{\lambda} = \sum_{\lambda_i=\lambda} \mathbf{u}_i.\mathbf{u}_i^t.$ If the downsampling function β is chosen as β_H or β_L , then the deformed eigenvector $\hat{\mathbf{u}}$ in Lemma 1 is equal to $\mathbf{J}_{\beta}\mathbf{u}$, which is an eigen-vector of \mathcal{L} with eigenvalue $2-\lambda$. It can also be seen that if eigenvectors $\{\mathbf{u}_i\}_{\lambda_i=\lambda}$ are orthogonal to each other then so are the deformed set of eigenvectors $\{\mathbf{J}_{\beta}\mathbf{u}_i\}_{\lambda_i=\lambda}$ and form basis of eigenspace $\mathbf{P}_{2-\lambda}$. Therefore, $\mathcal{L}\mathbf{J}_{\beta}\mathbf{P}_{\lambda}\mathbf{J}_{\beta} = \sum_{\lambda_i=\lambda} \mathcal{L}.\mathbf{J}_{\beta}\mathbf{u}_i.(\mathbf{J}_{\beta}\mathbf{u}_i)^t = \sum_{\lambda_i=\lambda} (2-\lambda).\mathbf{J}_{\beta}\mathbf{u}_i.(\mathbf{J}_{\beta}\mathbf{u}_i)^t = (2-\lambda)\mathbf{P}_{2-\lambda}$, therefore $\mathbf{J}_{\beta}\mathbf{P}_{\lambda}\mathbf{J}_{\beta} = \mathbf{P}_{2-\lambda}$ which implies that $\mathbf{J}_{\beta}\mathbf{P}_{\lambda} = \mathbf{P}_{2-\lambda}\mathbf{J}_{\beta}$.

We term this phenomenon, spectrum folding in bipartite graphs, as the deformed eigenvector (or eigenspace) for any $\lambda \in \sigma(G)$ appears as another eigenvector (or eigenspace) at a mirror eigenvalue around $\lambda=1$. To understand it, let ${\bf f}$ be an N-D graph-signal on bipartite graph G=(L,H,E) with eigenspace decomposition

$$\mathbf{f} = \sum_{\lambda \in \sigma(G)} \mathbf{P}_{\lambda} \mathbf{f} = \sum_{\lambda \in \sigma(G)} \mathbf{f}^{\lambda}, \tag{23}$$

where $\mathbf{f}^{\lambda} = \mathbf{P}_{\lambda}\mathbf{f}$ is the projection of \mathbf{f} onto the eigenspace V_{λ} and let the output signal after DU operation with down-sampling function β_L (or β_H) be \mathbf{f}_{du} . Then the V_{λ} eigenspace projection of the output signal is given as:

$$\mathbf{f}_{du}^{\lambda} = \mathbf{P}_{\lambda} \mathbf{f}_{du} = \frac{1}{2} \mathbf{P}_{\lambda} \mathbf{f} + \mathbf{P}_{\lambda} \mathbf{J}_{\beta_L} \mathbf{f}, \tag{24}$$

which using (22), can be written as:

$$\mathbf{f}_{du}^{\lambda} = \frac{1}{2} \mathbf{P}_{\lambda} \mathbf{f} + \mathbf{J}_{\beta_L} \mathbf{P}_{2-\lambda} \mathbf{f}$$
$$= \frac{1}{2} (\mathbf{f}^{\lambda} + \mathbf{J}_{\beta_L} \mathbf{f}^{2-\lambda}). \tag{25}$$

In (25), the distortion term $\mathbf{J}_{\beta_L}\mathbf{f}^{2-\lambda}$, which arises due to the downsampling of \mathbf{f}^{λ} has the same coefficients as that of $\mathbf{f}^{2-\lambda}$ (except for different signs). Further, the eigenspace decomposition of the output signal can be written as:

$$\mathbf{f}_{du} = \frac{1}{2} \sum_{\lambda \in \sigma(G)} (\mathbf{f}^{\lambda} + \mathbf{J}_{\beta} \mathbf{f}^{2-\lambda}) = \frac{1}{2} (\mathbf{f} + \mathbf{f}^{alias})$$
 (26)

In other words, the output signal is the average of the original signal and a shifted and aliased version of the original signal, and hence the term spectral folding. In the next Section, we utilize this property to design perfect reconstruction filterbanks for bipartite graphs.

B. Two-Channel Filterbank Conditions for Bipartite Graphs

Referring again to Figure 1, for bipartite graph G=(L,H,E), let $\beta_H=\beta$ be the downsampling function for \mathbf{H}_1 filter channel and $\beta_L=-\beta$ be the downsampling function for \mathbf{H}_0 channel. Thus the nodes in H only retain the output of highpass channel and nodes in L retain the output of the lowpass channel. In our proposed design, we also choose the synthesis filters \mathbf{G}_0 and \mathbf{G}_1 to be spectral filters with kernels $g_0(\lambda)$ and $g_1(\lambda)$ respectively 3 . Then, by using (5) and (21) the perfect reconstruction conditions in (13) can be rewritten as:

$$\mathbf{T}_{eq} = \mathbf{G}_{0}\mathbf{H}_{0} + \mathbf{G}_{1}\mathbf{H}_{1}$$

$$= \sum_{\lambda \in \sigma(G)} (g_{0}(\lambda)h_{0}(\lambda) + g_{1}(\lambda)h_{1}(\lambda)) \mathbf{P}_{\lambda}$$

$$\mathbf{T}_{alias} = \mathbf{G}_{1}\mathbf{J}_{\beta}\mathbf{H}_{1} - \mathbf{G}_{0}\mathbf{J}_{\beta}\mathbf{H}_{0}$$

$$= \sum_{\lambda,\gamma \in \sigma(G)} (g_{1}(\lambda)h_{1}(\gamma) - g_{0}(\lambda)h_{0}(\gamma)) \mathbf{P}_{\lambda}\mathbf{J}_{\beta}\mathbf{P}_{\gamma}.$$

$$(27)$$

1) Aliasing cancellation: Using (5) and the spectral folding property of bipartite graphs in (22), $T_{alias}f$ can be written as:

$$\mathbf{T}_{alias}\mathbf{f} = \sum_{\lambda \in \sigma(G)} (g_1(\lambda)h_1(2-\lambda) - g_0(\lambda)h_0(2-\lambda)) \,\mathbf{J}_{\beta}\mathbf{P}_{2-\lambda}\mathbf{f}$$
$$= \sum_{\lambda \in \sigma(G)} (g_1(\lambda)h_1(2-\lambda) - g_0(\lambda)h_0(2-\lambda)) \,\mathbf{P}_{\lambda}\mathbf{J}_{\beta}\mathbf{f}^{2-\lambda}$$

Since, $\mathbf{J}_{\beta}\mathbf{f}^{2-\lambda}$ is the aliasing term corresponding to \mathbf{f}^{λ} , $\mathbf{T}_{alias}\mathbf{f}$ is the aliasing part of the reconstructed signal, and an alias-free reconstruction using spectral filters is possible if and only if for all λ in $\sigma(G)$,

$$g_0(\lambda)h_0(2-\lambda) - g_1(\lambda)h_1(2-\lambda) = 0.$$
 (28)

2) Perfect reconstruction: Perfect reconstruction means that the reconstructed signal $\hat{\mathbf{f}}$ is the same as (or possibly a scaled version of) the input signal \mathbf{f} . $\mathbf{T}_{eq} + \mathbf{T}_{alias} = \mathbf{I}$. Therefore assuming the filterbanks cancel aliasing, the perfect reconstruction can be obtained if and only if $\mathbf{T}_{eq} = c^2\mathbf{I}$ for some scalar constant c. Thus, a necessary and sufficient condition for perfect reconstruction, using spectral filters, in bipartite graphs filterbanks is that for all λ in $\sigma(G)$,

$$g_0(\lambda)h_0(\lambda) + g_1(\lambda)h_1(\lambda) = c^2,$$

$$g_0(\lambda)h_0(2-\lambda) - g_1(\lambda)h_1(2-\lambda) = 0.$$
 (29)

3) Orthogonality: The equivalent analysis filter \mathbf{T}_a in the filterbank of Figure 1 is given as

$$\mathbf{T}_{a} = \frac{1}{2} \left((\mathbf{I} - \mathbf{J}_{\beta}) \mathbf{H}_{0} + (\mathbf{I} + \mathbf{J}_{\beta}) \mathbf{H}_{1} \right)$$
$$= \frac{1}{2} (\mathbf{H}_{0} + \mathbf{H}_{1}) + \frac{1}{2} \mathbf{J}_{\beta} (\mathbf{H}_{1} - \mathbf{H}_{0})$$
(30)

The filterbank provides an *orthogonal decomposition* of the graph signal if $\mathbf{T}_a^{-1} = \mathbf{T}_a^t$, which implies $\mathbf{T}_a \mathbf{T}_a^t = \mathbf{T}_a^t \mathbf{T}_a = \mathbf{I}$. Since, the spectral filters as well as the downsampling matrix \mathbf{J}_β are symmetric, $\mathbf{T}_a^t \mathbf{T}_a$ can be expanded as:

$$\mathbf{T}_a^t \mathbf{T}_a = \frac{1}{2} \left(\mathbf{H}_0^2 + \mathbf{H}_1^2 + \mathbf{H}_1 \mathbf{J}_\beta \mathbf{H}_1 - \mathbf{H}_0 \mathbf{J}_\beta \mathbf{H}_0 \right)$$
(31)

Combining (21) and (31) we obtain:

$$\mathbf{T}_{a}^{t}\mathbf{T}_{a} = 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda}$$

$$+ 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda}$$

$$+ 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda}$$

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$$+ 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda}$$

$$+ 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda} \mathbf{P}_{\lambda}$$

$$+ 1/2 \sum_{\lambda \in \sigma(G)} \underbrace{\left(h_{0}^{2}(\lambda) + h_{1}^{2}(\lambda)\right)}_{C_{\lambda}} \mathbf{P}_{\lambda} \mathbf{P}$$

+
$$1/2 \sum_{\lambda \in \sigma(G)} \underbrace{(h_1(\lambda)h_1(2-\lambda) - h_0(\lambda)h_0(2-\lambda))}_{D_{\lambda}} \mathbf{J}_{\beta} \mathbf{P}_{\lambda}$$

Thus, orthogonality can be obtained if and only if $C_{\lambda}\mathbf{I} + D_{\lambda}\mathbf{J}_{\beta} = c^2\mathbf{I}$ for some constant c and for all $\lambda \in \sigma(G)$, which is possible if and only if $D_{\lambda} = 0$ and $C_{\lambda} = c^2$ for all λ . Thus, a necessary and sufficient condition for orthogonality in bipartite graph filterbanks using spectral filters is:

$$h_0(\lambda)h_0(2-\lambda) - h_1(\lambda)h_1(2-\lambda) = 0$$

 $h_0^2(\lambda) + h_1^2(\lambda) = c^2.$ (33)

Note that, comparing (29) and (33), the orthogonality conditions can be obtained from the perfect reconstruction conditions by selecting $g_0(\lambda) = h_0(\lambda)$ and $g_1(\lambda) = h_1(\lambda)$. This is analogous to the case of standard filterbanks and leads to our proposed graph-QMF design as explained in the next Section.

C. Proposed Solution: Graph-QMF Design

We extend the well-known quadrature mirror filter (QMF) solution to the case of bipartite graphs. Our proposed solution, termed as graph-QMF, leads to the design of a single spectral kernel $h_0(\lambda)$ by selecting the other spectral kernels as:

$$h_1(\lambda) = h_0(2 - \lambda)$$

$$g_0(\lambda) = h_0(\lambda)$$

$$g_1(\lambda) = h_1(\lambda) = h_0(2 - \lambda)$$
(34)

Proposition 2 (QMF Filters on Graph): For a bipartite graph G=(L,H,E), let a two-channel filterbank be as shown in Figure 1 with the downsampling function $\beta=\beta_H$ and with spectral filters $\{\mathbf{H}_0,\mathbf{H}_1,\mathbf{G}_0,\mathbf{G}_1\}$ corresponding to spectral kernels $\{h_0(\lambda),h_1(\lambda),g_0(\lambda),g_1(\lambda)\}$ respectively. Then for any arbitrary choice of kernel $h_0(\lambda)$, the proposed graph-QMF solution cancels aliasing in the filterbank. In addition for $h_0(\lambda)^2+h_0(2-\lambda)^2=c^2$ for all $\lambda\in\sigma(G)$ and $c\neq 0$ the filterbank provides perfect reconstruction and an orthogonal decomposition of graph-signals.

Proof: Substituting (34) into (28) leads to $g_0(\lambda)h_0(2-\lambda) - g_1(\lambda)h_1(2-\lambda) = 0$ and aliasing is indeed canceled. The reconstructed signal $\hat{\mathbf{x}}$ in this case is simply equal to $(1/2)\mathbf{T}_{eq}\mathbf{x}$ and can be written as:

$$\hat{\mathbf{x}} = \frac{1}{2} \sum_{\lambda \in \sigma(G)} (h^2(\lambda) + h^2(2 - \lambda)) \mathbf{x}^{\lambda}$$
 (35)

Thus for $(h^2(\lambda)+h^2(2-\lambda))=c^2$ and $c\neq 0$, the reconstructed signal $\hat{\mathbf{x}}=\frac{c^2}{2}\mathbf{x}$ is a scaled version of original signal. Similarly applying the mirror design $h_1(\lambda)=h_0(2-\lambda)$ in the conditions (33) we get $h_0(\lambda)h_0(2-\lambda)-h_1(\lambda)h_1(2-\lambda)=0$ and $h_0^2(\lambda)+h_1^2(\lambda)=c^2$ and hence corresponding analysis side transform \mathbf{T}_a is orthogonal.

We now consider the design of kernels $h_0(\lambda)$ satisfying the design constraint of Proposition 2, i.e., for which $h_0^2(\lambda)$ +

³In general, synthesis filters do not have to be based on the spectral design. A case is presented in our previous work [18] with linear kernel spectral analysis filters and non-spectral synthesis filters.

 $h_0^2(2-\lambda)=c^2$ for all $\lambda\in\sigma(G)$. For maximum spectrum splitting in the two channels of the filterbank, the ideal choice of kernel $h_0(\lambda)$ would be a lowpass rectangular function on λ given as:

$$h_0^{ideal}(\lambda) = \begin{cases} c & \text{if } \lambda < 1\\ c/\sqrt{(2)} & \text{if } \lambda = 1\\ 0 & \text{if } \lambda > 1 \end{cases}$$
 (36)

The corresponding ideal filter is given by

$$\mathbf{H}_{0}^{ideal} = \sum_{\lambda < 1} c \mathbf{P}_{\lambda} + \frac{c}{\sqrt{2}} \mathbf{P}_{\lambda = 1}$$
 (37)

Note that the ideal transform has a non-analytic spectral kernel response with sharp peaks and is therefore a global transform (i.e., the filter operations are not localized). Even analytic solutions of the constraint equation $h_0^2(\lambda) + h_0^2(2-\lambda) = c^2$, such as $h_0(\lambda) = c\sqrt{1-\lambda/2}$ or $h_0(\lambda) = c\cos(\pi\lambda/4)$, are not very well localized in the spatial domain. By relaxing the constraints one can obtain spatially localized solutions at the cost of some small reconstruction error and nearperfect orthogonality. One such solution is the approximation of the desired kernel with a polynomial kernel. We choose polynomial approximations of the desired kernel due to the following localization property for corresponding transforms:

Lemma 2 ([8]): Let $h_0(\lambda)$ be a polynomial of degree k and let \mathcal{L} be the normalized Laplacian matrix for any weighted graph G, then the matrix polynomial $\mathbf{H}_0 = h_0(\mathcal{L})$ is exactly k-hop localized at each node of G. In other words for any two nodes n and m if $m \notin \mathcal{N}_k(n)$ then $\mathbf{H}_0(n,m) = 0$.

Further, we choose a minimax polynomial approximation which minimizes the Chebychev norm (worst-case norm) of the reconstruction error since it has been shown in [8] that it also minimizes the upper-bound on the error $||H^{ideal}-H^{poly}||$ between ideal and approximated filters. Thus, in order to localize the filters on the graph, we approximate h_0^{ideal} with the truncated Chebychev polynomials (which are a good approximation of minimax polynomials) of different orders. However since h_0^{ideal} is a rectangular function it projects a lot of its energy in the truncated part of the polynomial expansions and as a result the polynomial approximation errors for h_0^{ideal} are high. A possible solution of this problem is to soften the ideal case, by finding a smooth function that is low-pass and satisfies the constraint. An analogous construction in regular signal processing is Meyer's wavelet design which replaces the brick-wall type ideal frequency-response with a smooth scaling function that satisfies the orthogonality and scaling requirements. By a change in variable from $\omega \in [-1,1]$ to $\lambda \in [0,2]$ we can extend Meyer's wavelet construction in the case of bipartite graph. The construction involves choosing a function $\nu(x)$ such that $\nu(\lambda) = 0$ for $\lambda \leq 0$, $\nu(\lambda) = 1$ for $\lambda \geq 1$ and $\nu(\lambda) + \nu(1 - \lambda) = 1$ everywhere. One such function is given as:

$$\nu(\lambda) = \begin{cases} 0 & \text{if } \lambda \le 0\\ 3\lambda^2 - 2\lambda^3 & \text{if } 0 \le \lambda \le 1\\ 1 & \text{if } \lambda \ge 1 \end{cases}$$
 (38)

The smooth kernel is then given as:

$$h_0^{Meyer}(\lambda) = \sqrt{\nu(2 - \frac{3}{2}\lambda)}$$
 (39)

In Figure 2(a), we plot the ideal and Meyer wavelet kernels and in Figures 2(b)-(f) we plot the reconstruction errors between desired kernels and their polynomial approximations of different orders. It can be seen that Meyer's wavelet approximations

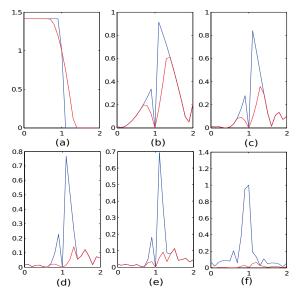


Fig. 2: (a) Ideal kernel (blue) vs. Meyer's wavelet kernel (red). It can be seen that Meyer's wavelet has smoother transition at $\lambda=1$ than the ideal kernel, (b)-(f) the reconstruction error magnitudes between original kernels and their polynomial approximations of order 2,4,6,8 and 10 respectively: ideal kernel (blue curves) and Meyers kernel(red curve).

yield small reconstruction errors as compared to ideal-filter approximations. Thus by choosing $h(\lambda)$ as the low-order polynomial approximations of smooth low-pass functions (such as Meyer's wavelets), we obtain near perfect reconstruction QMF wavelet filters on any bipartite graph which are very well localized in spatial domain.

D. Multi-dimensional separable wavelet filterbanks for arbitrary graphs

Not all graphs are bipartite. In order to apply our filterbank design to an arbitrary graph, $G = (\mathcal{V}, E)$, we propose a separable downsampling and filtering approach, where our previously designed two-channel filterbanks are applied in a "cascaded" manner, by filtering along a series of bipartite subgraphs of the original graph. This is illustrated in Figure 3. We call this a "separable" approach in analogy to separable transforms for regular multidimensional signals. For example in the case of separable transforms for 2D signals, filtering in one dimension (e.g., row-wise) is followed by filtering of the outputs along the second dimension (column-wise). In our proposed approach, a stage of filtering along one "dimension" corresponds to filtering using only those edges that belong to the corresponding bipartite subgraph. As shown in Figure 3, after filtering along one subgraph the results are stored in the vertices, and a new transform is applied to the resulting graph signals following the edges of the next level bipartite subgraph.

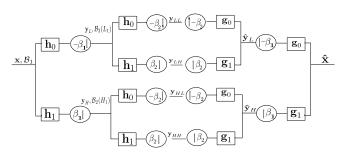


Fig. 3: Block diagram of a 2D Separable two-channel Filter Bank: the graph G is first decomposed into two bipartite subgraphs \mathcal{B}_1 and \mathcal{B}_2 , using the proposed decomposition scheme. By construction \mathcal{B}_2 is composed of two disjoint graphs $\mathcal{B}_2(L)$ and $\mathcal{B}_2(H)$, each of which is processed independently, by one of the two filterbanks at the second stage. The 4 sets of output transform coefficients, denoted as $\mathbf{y}_{HH}, \mathbf{y}_{HL}, \mathbf{y}_{LH}$ and \mathbf{y}_{LL} , are stored at disjoint sets of nodes.

In what follows we will assume that G has been decomposed into a series of K bipartite subgraphs $\mathcal{B}_i = (L_i, H_i, E_i)$, $i = 1 \dots K$; how such a decomposition may be obtained will be discussed later. The bipartite subgraphs cover the same vertex set: $L_i \cup H_i = \mathcal{V}, i = 1, 2, ...K$. Each edge in G belongs to exactly one E_i , i.e., $E_i \cap E_j = \emptyset$, $i \neq j$, $\bigcup_i E_i = E$. Note that in each bipartition we need to decide both a 2coloring (H_i, L_i) and an assignment of edges (E_i) . In order to guarantee invertibility for structures such as those of Figure 3, given the chosen 2-colorings (H_i, L_i) , the edge assignment has to be performed iteratively based on the order of the subgraphs. That is, edges for subgraph 1 are chosen first, then those for subgraph 2 are selected, and so on. The basic idea is that at each stage i all edges between vertices of different colors that have not been assigned yet will be included in E_i . More formally, at stage i with sets H_i and L_i , E_i contains all the links in $E - \bigcup_{k=1}^{i-1} E_k$ that connect vertices in L_i to vertices in H_i . Thus E_1 will contain all edges between H_1 and L_1 . Then, we will assign to E_2 all the links between nodes in H_2 and L_2 that were not already in E_1 . This is also illustrated in Figure 4. Note that, by construction $G_1 = G - \mathcal{B}_1 = (\mathcal{V}, E - E_1)$ contains now two disjoint graphs, since all edges between L_1 and H_1 were assigned to E_1 . Thus, at the second stage in Figure 3, \mathcal{B}_2 is composed of two disjoint graphs $\mathcal{B}_2(L_1)$ and $\mathcal{B}_2(H_1)$, which each will be processed independently by one of the two filterbanks at this second stage. Clearly, this guarantees invertibility of the decomposition of Figure 3, since it will be possible to recover the signals in $\mathcal{B}_2(L_1)$ and $\mathcal{B}_2(H_1)$ from the outputs of the 2nd stage of the decomposition. The same argument can be applied to the decompositions with more than two stages. That is, the output of a two-channel filterbank at level i leads to two subgraphs, one per channel, that are disconnected when considering the remaining edges $(E - \bigcup_{k=1}^{i} E_k)$. The output of a K-level decomposition leads to 2^K disconnected subgraphs.

We now derive expressions for the proposed cascaded transform along bipartite subgraphs. Using K=2 case as an example, assuming that the original graph can be approximated exactly with two bipartite subgraphs as shown in Figure 4, we choose $\beta_i = \beta_{H_i}$ as the downsampling function for bipartite graph \mathcal{B}_i , for i=1,2. Further, let us denote \mathbf{J}_{β_i} ,

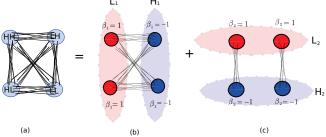


Fig. 4: Example of 2-dimensional separable downsampling on a graph: (a) original graph G, (b) the first bipartite graph $\mathcal{B}_1=(L_1,H_1,E_I)$, containing all the links in G between sets L_1 and H_1 . (c) the second bipartite graph $\mathcal{B}_2=(L_2,H_2,E_2)$, containing all the links in $G-\mathcal{B}_1$, between sets L_2 and H_2

as the downsampling matrices, and \mathbf{H}_{i0} and \mathbf{H}_{i1} as the lowpass and high-pass graph-QMF filters respectively, for the bipartite graph \mathcal{B}_i , for i=1,2. Since, the vertex sets L_1 and H_1 in bipartite graph \mathcal{B}_2 are disconnected, the filtering and downsampling operations on graphs $\mathcal{B}_2(L_1)$ and $\mathcal{B}_2(H_1)$ do not interact with each other. Therefore, graph-filters \mathbf{H}_{2j} , for j=0,1 on the second bipartite graph \mathcal{B}_2 , can be represented as *block-diagonal matrices* with diagonal entries $\mathbf{H}_{2j}(H_1,H_1)$ and $\mathbf{H}_{2j}(L_1,L_1)$. As a result, \mathbf{H}_{20} and \mathbf{H}_{21} commute with downsampling matrix \mathbf{J}_{β_1} of the first bipartite subgraph, i.e.,

$$\mathbf{H}_{2j}\mathbf{J}_{\beta_1} = \mathbf{J}_{\beta_1}\mathbf{H}_{2j},\tag{40}$$

for j=1,2. ³ Further, let \mathbf{T}_{ai} be the equivalent analysis transform for \mathcal{B}_i , for i=1,2. The combined analysis transform \mathbf{T}_a in the 2-dimensions can be written as the product of analysis transform in each dimension. Using (30), we obtain:

$$\mathbf{T}_{a} = \mathbf{T}_{a2}.\mathbf{T}_{a1} = \prod_{i=1}^{2} \frac{1}{2} \left((\mathbf{H}_{i1} + \mathbf{H}_{i0}) + \mathbf{J}_{\beta_{i}} (\mathbf{H}_{i1} - \mathbf{H}_{i0}) \right),$$
 (41)

Note that, for exact graph-QMF filter design such as with the Meyer kernel in (39), \mathbf{T}_{ai} is invertible with $\mathbf{T}_{ai}^{-1} = \mathbf{T}_{ai}^t$, for i=0,1. As a result, \mathbf{T}_a is invertible with $\mathbf{T}_a^{-1} = \mathbf{T}_{a1}^t.\mathbf{T}_{a2}^{t-2}$. The transform function \mathbf{T}_a can be further decomposed into the transform functions \mathbf{T}_{HH} , \mathbf{T}_{Hl} , \mathbf{T}_{LH} and \mathbf{T}_{LL} corresponding to the four channels in Figure 3. For example, the transform \mathbf{T}_{HH} , consists of all the terms in the expansion of \mathbf{T}_a in (41), containing filters H_{11} and H_{21} . Thus,

$$\mathbf{T}_{HH} = \frac{1}{4} (\mathbf{H}_{21} \mathbf{H}_{11} + \mathbf{H}_{21} \mathbf{J}_{\beta_1} \mathbf{H}_{11} + \mathbf{J}_{\beta_2} \mathbf{H}_{21} \mathbf{H}_{11} + \mathbf{J}_{\beta_2} \mathbf{H}_{21} \mathbf{J}_{\beta_1} \mathbf{H}_{11}),$$
(42)

where $(1/4)\mathbf{H}_{21}\mathbf{H}_{11}$ is the transform without downsampling, and the remaining terms arise primarily due to the downsampling in the HH channel. Using (40), which is a property of our proposed decomposition scheme in (42), we obtain:

$$\mathbf{T}_{HH} = \frac{1}{4} (\mathbf{H}_{21} \mathbf{H}_{11} + \mathbf{J}_{\beta_1} \mathbf{H}_{21} \mathbf{H}_{11} + \mathbf{J}_{\beta_2} \mathbf{J}_{\beta_1} \mathbf{H}_{21} \mathbf{H}_{11})$$

$$= \frac{1}{4} (\mathbf{I} + \mathbf{J}_{\beta_2}) (\mathbf{I} + \mathbf{J}_{\beta_1}) \mathbf{H}_{21} \mathbf{H}_{11}.$$
(43)

³In general, this result can be applied to any general K-dimensional decomposition using proposed recursive method, as the downsampling matrix \mathbf{J}_{β_i} commutes with all filter matrices \mathbf{H}_{k1} and \mathbf{H}_{k2} corresponding to bipartite subgraph \mathcal{B}_k , where k > i.

⁴For polynomial approximations, of Meyer kernels, we incur some reconstruction errors in each dimension.

Thus, the equivalent transform in each channel of the proposed 2-dimensional separable filterbanks can be interpreted as filtering with a 2-dimensional filter, such as $H_{21}H_{11}$ for the HH channel, followed by DU operations with two downsampling functions $\beta_2(n)$ and $\beta_1(n)$ in cascade. It also follows from (44), that the output of $\mathbf{H}_{21}\mathbf{H}_{11}$ in the HHchannel is stored only at the nodes corresponding to $H_1 \cap H_2$. Thus, the output of each channel is stored at mutually disjoint sets of nodes, and each node stores the output of exactly one of the channel. Therefore, the overall filterbank is critically sampled. Further, if the spectral decompositions of \mathcal{B}_1 and \mathcal{B}_2 are given as $\{\lambda, \mathbf{P}_{\lambda}^1\}$ and $\{\gamma, \mathbf{P}_{\gamma}^2\}$, then $\mathbf{H}_{21}\mathbf{H}_{11}$ consists of a two dimensional spectral kernel $h_{21}(\gamma)h_{11}(\lambda)$ and corresponding eigenspace $\mathbf{P}^{2}_{\lambda}\mathbf{P}^{1}_{\lambda}$. The analysis extends to any dimension K > 2 with K-dimensional graph-frequencies $(\lambda_1, \lambda_2, ..., \lambda_K)$, corresponding eigenspace $\mathbf{P}_{\lambda_1}^1, \mathbf{P}_{\lambda_2}^2, ... \mathbf{P}_{\lambda_K}^K$ and transforms with spectral response $\prod_{i=1}^{K} g_i(\lambda_i)$. So far we have described, how to implement separable

So far we have described, how to implement separable multi-dimensional graph-QMF filterbanks on a graph G, given a decomposition of G into K bipartite subgraphs. In particular, we defined a "separable" method of graph decomposition, which leads to a cascaded tree-structured implementation of the multi-dimensional filterbanks. While these multi-dimensional filterbanks can be implemented for *any* separable bipartite subgraph decomposition of G, the definition of a "good" bipartite decomposition of any arbitrary graph remains a topic for future work, and may be application dependent. In this paper, we propose a bipartite subgraph decomposition method, referred to as *Harary's decomposition*, which provides a $\lceil log_2k \rceil$ bipartite decomposition of a graph G given a k-coloring defined on it. The method is derived from $\lceil 19 \rceil$ and we describe it in Algorithm 1.6

Note that invertible cascaded transforms can also be constructed even when the conditions for edge selection described are not followed, e.g., if an edge e_1 between nodes in H_1 and L_1 is not included in E_1 . In such a situation, it is possible to perform an invertible cascaded decomposition if e_1 is no longer used in further stages of decomposition. Thus, we would have an invertible decomposition but on a graph that approximates the original one (i.e., without considering e_1). Alternatively it can be shown that it is possible to design invertible transforms with arbitrary E_i selections (i.e., not following the rules set out in this paper), but these transforms are not necessarily critically sampled. A more detailed study of this case falls outside of the scope of this paper.

E. Multiresolution decomposition using two-channel filterbanks

The two-channel filterbanks on a single bipartite graph $\mathcal{B} = (H, L, E)$ have the property of decomposing the signal into two lower-resolution versions $\hat{\mathbf{f}}_L$ and $\hat{\mathbf{f}}_H$ respectively, as in (11). The signal $\hat{\mathbf{f}}_L$ is a lowpass or coarse resolution

Algorithm 1 Harary's Decomposition

Require: F, s.t. F(v) is the color assigned to node v, min(F)=1, max(F)=k.

- 1: Set L_1 = set of nodes with $F(v) \leq \lfloor k/2 \rfloor$ colors.
- 2: Set H_1 = set of nodes with $F(v) > \lceil k/2 \rceil$ colors.
- 3: Set $E_1 \subset E$ containing all the edges between sets H_1 and L_1 .
- 4: Compute bipartite subgraph $\mathcal{B}_1 = (L_1, H_1, E_1)$,
- 5: Set $G = G \mathcal{B}_1$.
- 6: G is now a union of two disconnected subgraphs $G(H_1)$ and $G(L_1)$.
- 7: Graph $G(L_1)$ is $\lceil k/2 \rceil$ -colorable.
- 8: Compute coloring \mathbf{F}_L on $G(L_1)$ s.t. $\min(F_L)=1$, $\max(F_L)=\lceil k/2 \rceil$.
- 9: Graph $G(H_1)$ is $\lfloor k/2 \rfloor$ -colorable.
- 10: Compute coloring \mathbf{F}_H on $G(H_1)$ s.t. $\min(F_H)=1$, $\max(F_L)=|k/2|$.
- 11: Repeat 1-4 on $G(L_1)$ and $G(H_1)$ to obtain bipartite subgraphs $\mathcal{B}_2(L_1)$ and $\mathcal{B}_2(H_1)$.
- 12: Compute bipartite subgraph $\mathcal{B}_2 = \mathcal{B}_2(L_1) \cup \mathcal{B}_2(H_1)$.
- 13: Set $G = G \mathcal{B}_2$.
- 14: repeat 1-13 exactly $\lceil log_2k \rceil$ times after which graph G will become an empty graph.

version constructed from the output coefficients of the lowpass channel stored on the set L, whereas f_H is a highpass version of the input constructed from the output coefficients of the filterbank stored on the set H. Analogous to treestructured filterbanks for 1-D signals, this decomposition can be applied recursively on the low-pass (or high-pass) signal by constructing a downsampled graph consisting of vertices in L (or H) and some appropriate edge-structure. One way to compute the downsampled graph G_L (or G_H) is to reconnect two nodes in set L (or H) if they are 2-hops away in the original graph. Note that for bipartite graphs, unlike the case of regular lattices, the resulting downsampled graphs G_L and G_H may neither be identical nor bipartite. Therefore, for the next level of decomposition, we can either operate on a single bipartite graph approximation of G_L which leads to a onedimensional two-channel filterbank, or a multiple bipartite graph approximation, which leads to a multi-dimensional two-channel filterbank implementation on the downsampled graph. Further, this multiresolution decomposition of graphsignals can be extended to the case of general K-dimensional two-channel filterbanks for any arbitrary graph G, which decomposes the signal into 2^K lower-resolution versions, as described in Section III-D. In this case, the downsampled graphs in each channel, can be computed by reconnecting two nodes in the downsampled vertex set, if they are 2^K -hops away in the original graph.

IV. EXPERIMENTS

A. Graph-QMF Design Details

We first provide explicit details of the filterbank design for arbitrary graphs. Given any arbitrary undirected graph $G=(\mathcal{V},E)$, we find a minimum perfect-coloring χ of its

 $^{^5}$ A graph is perfectly k-colorable if its vertices can be assigned k-colors in such a way that no two adjacent vertices share the same color. The term *chromatic number* $\chi(G)$ of a graph refers to smallest such k.

⁶Note that the bipartite decomposition is not unique and depends on the ordering in which the k-colors are divided.

vertices using a graph-coloring algorithm, such as the BSC algorithm given in [20]. The coloring information is then used to decompose G into a set of $K = \lceil log_2(\chi) \rceil$ bipartite graphs $\mathcal{B}_i = (L_i, H_i, E_i)$, for i = 1, 2, ...K using Harary's Algorithm as described in Section III-D. For each subgraph \mathcal{B}_i , we compute its normalized Laplacian matrix \mathcal{L}_i and the downsampling function $\beta_i = \beta_{H_i}$. Further, we compute the low-pass analysis kernel $h_{i,0}(\lambda)$ on \mathcal{B}_i , as the $m_i^{\hat{t}h}$ order Chebychev approximation of the Meyer kernel $h_0^{Meyer}(\lambda)$, for some positive integer value m_i . The remaining spectral kernels $h_{i,1}(\lambda), q_{i,0}(\lambda), q_{i,1}(\lambda)$ are computed from $h_{i,0}(\lambda)$ according to graph-QMF relations mentioned in (34). The corresponding analysis and synthesis transforms are then computed as $\mathbf{H}_{i,j} =$ $h_i(\mathcal{L}_i)$ and $G_{i,j} = g_i(\mathcal{L}_i)$, respectively, for j = 0, 1. Note that, since the kernels are polynomials, the transforms are also matrix polynomials of Laplacian matrices and do not require explicit eigenspace decompositions. In our experiments, we use $m_i = m$, and hence $h_{i,j}(\lambda) = h_j(\lambda), j \in \{0,1\}$ for all i, in which case the resulting transforms are exactly mhop localized on each bipartite subgraph. The order m is a parameter of our design and should be chosen based on the required level of spatial localization and how much reconstruction error can be tolerated. The overall filterbank is designed by concatenating filterbanks of each bipartite subgraph in the form of a tree, analogous to Figure 3 in the 2-dimensional decomposition case. We now describe some experiments to demonstrate potential applications of our proposed filterbanks.

B. Graph Filter-banks on Images

Digital images are 2-D regular signals, but they can also be formulated as graphs by connecting every pixel (node) in an image with its neighboring pixels (nodes) and by interpreting pixel values as the values of the graph-signal at each node. The graph-representations of the regular-signals are shown to be promising in practice recently [21], [22]. Figure 5 shows some of the ways in which pixels in an image can be connected with other pixels to formulate a graph representation of any image. The advantage of using a graph formulation of the images is that it provides flexibility of linking pixels in arbitrary ways, leading to different filtering/downsampling patterns. To demonstrate this, we implement an ideal spectral

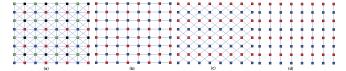


Fig. 5: Graph-formulation of 2D images: (a) 8-connected image-graph G, which is 4-partite. The node partitions are represented by colors of the nodes. The graph G can be decomposed into (b) a 4-connected rectangular image-graph G^r , and (c) a 4-connected diagonal image graph G^d both of which are 2-partite. (d) a 2-connected image-grapp G_v with vertical links only.

low-pass filters on the graph formulations of the 2D images, shown in Figure 5. Since, the graphs G^r , G^d , and G^v are bipartite graphs, the ideal spectral lowpass filter \mathbf{H}_0^{ideal} on these graph can be computed as in (37). In Figure 6, we plot the DFT magnitude response of ideal lowpass spectral

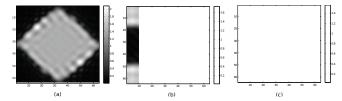


Fig. 6: DFT magnitude responses of the ideal spectral filters \mathbf{H}_0^{ideal} on (a) G_r , (b) G_d and (c) G_v .

transforms on these bipartite image-graphs. 5 We observe in Figure 5(b) that, the downsampling pattern (red/blue nodes) on the rectangular subgraph G^r is identical to the quincunx downsampling pattern, and in Figure 6(a), it can be observed that the DFT magnitude response of \mathbf{H}_0^{ideal} filter on G^r is same as the DFT magnitude response of the standard anti-aliasing filter for quincunx downsampling. Similarly, we observe that the spectral low-pass filter for G^v in Figure 5(d) has the same DFT magnitude responses (Figure 6(c)) as the anti-aliasing filters for vertical factor-of-2 downsampling case. The graph formulation of images also allows us to explore new downsampling patters, for example, the image pixels can be connected to their diagonally opposite neighbors as shown in Figure 5(c). The DFT magnitude response of the ideal spectral low-pass filter in this case, is shown in Figure 6(b) and has a wider passband in the diagonal directions. Further, in the nonbipartite graph formulation of the anti-aliasing filter for any arbitrary graph is the product of ideal-lowpass filters along its bipartite subgraph decompositions. Therefore the rectangular graph G^r can be further decomposed into bipartite subgraph G^v and G^h leading to a rectangular (factor of 4) downsampling pattern.

This graph-based approach also provides additional degrees of freedom (directions) to filter/downsample the image while still having a critically sampled output. To demonstrate this, we implement a graph wavelet filterbank on the 8-connected image-graph G of a given image. The chromaticity of G is $\chi = 4$ (represented as different shape nodes in Figure 5(a)) and hence it can be decomposed into two edge-disjoint bipartite subgraphs. Among several such possible decompositions, we choose the decomposition that gives us a rectangular subgraph G^r and a diagonally connected diamond graph G^d . On each subgraph we implement a graph-QMF filterbank, as described in Section IV-A above. The resulting 2-dim separable filterbank has four channels as shown in Figure 3 and the nodes representing a specific shape in Figure 5(a) store the output of a specific channel.⁶ Figure 7 shows the output wavelet coefficients of proposed 2-dim filterbank on a toy image which has both diagonal and rectangular edges. In Figure 7, the energy of wavelet coefficients in the LH channel (lowpass on G^r , high-pass on G^d) is high around the rectangular

⁵Because of the regularity and symmetry of the links, the resulting filters at each node, are translated version of each other (except at the boundary nodes), and so we can compute the 2-D DFT magnitude response of a spectral transform, by computing the DFT response of the filtering operations at a single node.

 $^{^6}$ In general for an arbitrary graph with K-proper colors, the bipartite decomposition provides exactly K non-empty channels and nodes of a particular color store the output of a particular channel.

edges, which is reasonable, since subgraph G^d is diagonally connected and its low-pass spectral frequencies are oriented along diagonal links. Similarly we observe that the highenergy wavelet coefficients in the HL channel (high-pass on G^r , low-pass on G^d) lie around the diagonal edges, since G^r is rectangularly connected and its low-pass spectral frequencies are oriented towards horizontal and vertical directions. This example also shows that the filterbanks based on only NWSE connectivity are more suited for images with horizontal and vertical edges whereas the transform based only on diamond connectivity are more suited for image with diagonal edges. In Figure 8, we show the graph-wavelet decomposition of a depth-map image taken from [23]. Again, we see that the LH channel has high energy coefficients along nearly rectangular edges while the HL channel has high energy coefficients along nearly diagonal directions. More directions can be added to downsample/filter by increasing the connectivity of the pixels in the image-graph. Moreover, since graph-based transforms operate only over the links between nodes, the graph formulation is useful in designing edge-aware transforms, such as [21], [22], (which avoid filtering across edges) by removing links between pixels across edges.

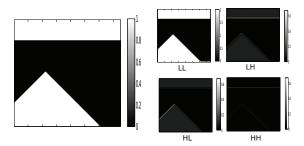


Fig. 7: Separable two-dim two channel graph filterbank on a toy image with both rectangular and diagonal edges.

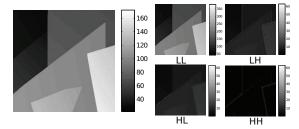


Fig. 8: Separable two-dim two channel graph filterbank on a depth-map image with parameter m=6

C. Graph Filter-banks on Irregular Graphs

Our proposed filterbanks can be used as a useful tool in analyzing/compressing arbitrarily linked irregular graphs. In order to demonstrate it, we take the example of *Minnesota traffic graph* from [8]. The graph is shown in Figure 9(a), where the spatial coordinates are only used to display the graph and the wavelet transform, and do not affect the edgeweights. Further, we consider a graph-signal on this graph with sharp irregular discontinuity, as shown in Figure 9(b), where the color of a node represents the signal value at that

node. The graph is perfectly 3-colorable and hence, we can decompose it into $\lceil log_2(3) \rceil = 2$ bipartite subgraphs, which are shown in Figure 9(c-d). Given such a decomposition,

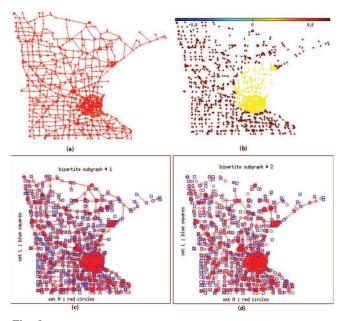


Fig. 9: (a) The Minnesota traffic graph G, and (b) the graph-signal to be analyzed. The colors of the nodes represent the sample values. (c)(d) bipartite decomposition of G into two bipartite subgraphs using Harary's decomposition.

we implement a two-channel graph-QMF filterbank on each bipartite subgraph, leading to a two-dimensional separable filterbank implementation, as shown in Figure 3. Since, the proper coloring of the Minnesota graph is 3, the output of the HL channel (i.e. nodes for which $(\beta_1(n), \beta_2(n)) = (-1, 1)$) is empty after downsampling. The output coefficients of the other three non-empty channels (LL, LH, HH) are shown in Figure 10. Note that, after downsampling, the total number of

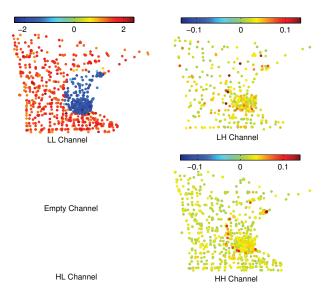


Fig. 10: Output coefficients of the proposed filterbanks with parameter m=24. The node-color reflects the value of the coefficients at that point. Top-left: LL channel wavelet coefficients, top-right: absolute value of LH channel wavelet coefficients, and bottom-right: absolute value of HH channel wavelet coefficients

output coefficients in the four channels is equal to the number of input samples, thus making the transform critically sampled. We observe in Figure 10 that the output coefficients in the LH and HH channels, have significantly high magnitude along the discontinuity, hence reflecting the high-pass nature of these channels. Further, in order to see how much energy of the original signal is captured in each channel, we upsample then filter the coefficient of each channel by the synthesis part of proposed filterbank. This is shown in Figure 11. In this figure, we see that the reconstructed signal from LLchannel coefficients, provides a low-pass approximation of the original signal (sharp boundaries blurred), whereas the signals reconstructed from the LH and HL channels provide a highpass approximation of the input signal (highlighting the boundaries). Thus, the proposed graph-based filterbanks, provide a meaningful decomposition of input signals, analogous to the standard wavelet-filterbanks.

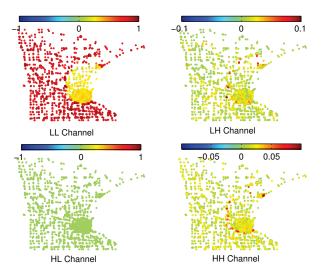


Fig. 11: Reconstructed graph-signals from the wavelet coefficients of individual channels. As before the node-color reflects the value of the coefficients at that node. Top-left: reconstruction from LL channel only, top-right: reconstruction from LH channel only, and bottom-right; reconstruction from HH channel only, Since, HL channel is empty the reconstruction is an all-zero signal (bottom-left figure).

V. CONCLUSION AND FUTURE WORK

We have proposed the construction of critically sampled wavelet filterbanks for analyzing graph-signals defined on any arbitrary finite weighted graph. For this we have formulated a bipartite subgraph decomposition problem which produces an edge-disjoint collection of bipartite subgraphs. For these bipartite graphs we have described and proved a spectrum folding phenomenon which occurs in downsampling then upsampling (DU) operations and produces aliasing in the graph signals. Based on this result, we have proposed twochannel wavelet filterbanks on bipartite graphs and provided necessary and sufficient conditions for aliasing cancellation, perfect reconstruction and orthogonality in these filterbanks. As a practical solution, we have proposed a graph-QMF design for bipartite graphs which has all the above mentioned features. The filterbanks are however, realized by Chebychev polynomial approximations at the cost of small reconstruction error and loss of orthogonality. Our current efforts are focused on finding solutions other than the proposed graph-QMF design and to understand and differentiate 'good' and 'bad' decompositions of arbitrary graphs into bipartite subgraphs.

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