

DESIGN OF CONJUGATE QUADRATURE FILTERS HAVING SPECIFIED ZEROS

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ABSTRACT

Conjugate quadrature filters with multiple zeros at 1 have classical applications to unitary subband coding of signals using exact reconstruction filter banks. Recent work shows how to construct, given a set of n negative numbers, a CQF whose degree does not exceed $2n-1$ and whose zeros contain the specified negative numbers, and applies such filters to interpolatory subdivision and to wavelet construction in Sobolev spaces. This paper describes a recent result of the authors which extends this construction for an arbitrary set of n nonzero complex numbers that contains no negative or negative reciprocal conjugate pairs. Detailed derivations are to be given elsewhere. We design several filters using an exchange algorithm to illustrate a conjecture concerning the minimal degree and we discuss an application to coding transient acoustic signals.

1. INTRODUCTION

For convenience identify filters (finite sequences) with their z -transforms (Laurent polynomials) and let $\mathcal{L}_R, \mathcal{L}_N$ denote the set filters whose restriction to the unit circle $T := \{z \in C : |z| = 1\}$ is real, nonnegative, respectively. In this paper n denotes a positive integer, Λ denotes a set of n (counted with multiplicity) nonzero complex numbers, and $u(\Lambda)$ denotes the subset of the unit disc obtained by replacing the elements in Λ outside the unit disc by their reciprocal conjugates.

Mintzer [1], Smith and Barnwell [2], [3], and Vetterli [4] invented conjugate quadrature filters P , that satisfy

$$|P(z)|^2 + |P(-z)|^2 = 1, \quad z \in T, \quad (1)$$

to perform unitary (lossless) subband coding of discrete signals using exact reconstruction filter banks. Let $Q(\Lambda)$ denote the set of all CQF's whose zeros contain Λ . Recently Micchelli showed if Λ a subset of negative numbers, $Q(\Lambda)$ contains a filter having degree $\leq 2n - 1$ and applied these filters to interpolatory subdivision [5] and to wavelet construction in Sobolev spaces [6]. The construction of symmetric CQF's, necessarily having complex coefficients, were described by Lawton in [7], where their utility for handling boundaries in signal coding was demonstrated. In a recent paper [8] we proved:

Result 1 $Q(\Lambda) \neq \phi$ iff

$$(\Lambda \cup \bar{\Lambda}^{-1}) \cap -(\Lambda \cup \bar{\Lambda}^{-1}) = \phi. \quad (2)$$

The only if part follows directly from (1). The if part is proved by constructing a minimal degree $P \in Q(\Lambda)$ as follows: define $S_1(z) := \prod_{\lambda \in \Lambda} (z - \lambda)$, $z \in C$, and construct filters $P_1, P_2 \in \mathcal{L}_N$ by $P_1(z) := |S_1(z)|^2$, $z \in T$, and $P_2(z) := P_1(-z)$, $z \in C$. Clearly the pair P_1, P_2 satisfies the hypothesis of the following result proved in [8]

Result 2 If the pair $P_1, P_2 \in \mathcal{L}_N$ have no common zeros in $C \setminus \{0\}$, there exists a pair $Q_1, Q_2 \in \mathcal{L}_N$ such that

$$P_1(z)Q_1(z) + P_2(z)Q_2(z) = 1. \quad (3)$$

Choose a pair $Q_1, Q_2 \in \mathcal{L}_N$ whose coefficient sequences have minimal lengths and that satisfy (3). Define $W \in \mathcal{L}_N$ by $W(z) = 1/2(Q_1(z) + Q_2(-z))$. Then construct a spectral factor S_2 of W whose coefficient sequence is supported on the nonnegative integers including 0 and define $P := S_1 S_2$. Then (3) implies $P \in Q(\Lambda)$ and has minimal degree. The filter $G := P_1 W$ is *interpolatory* since $G(z) + G(-z) = 1$ and P is a spectral factor of G . Furthermore, if $\Lambda = \bar{\Lambda}$ then W will have real coefficients. Therefore P will have real coefficients if S_2 is chosen to have real coefficients. This will be the case if S_2 is the *minimal phase* (roots are in unit disc) spectral factor of W .

We describe only the basic concept of our proof of Result 2 given in [8]. First, we used standard algebraic methods to construct minimal length filters $B_1, B_2 \in \mathcal{L}_R$ such that

$$P_1 B_1 + P_2 B_2 = 1 \quad (4)$$

and showed that a pair of filters $Q_1, Q_2 \in \mathcal{L}_R$ satisfies (3) if and only if there exists $F \in \mathcal{L}_R$ such that

$$Q_1 = B_1 - F P_2; \quad Q_2 = B_2 + F P_1. \quad (5)$$

Second, we defined rational functions

$$R_1 := -\frac{B_2}{P_1}; \quad R_2 := \frac{B_1}{P_2} \quad (6)$$

and showed Q_1 and Q_2 defined by (5) are in \mathcal{L}_N if and only if

$$R_1(z) \leq F(z) \leq R_2(z), \quad z \in T. \quad (7)$$

Third, we used approximation theoretic methods to show the existence of $F \in \mathcal{L}_R$ that satisfies (7).

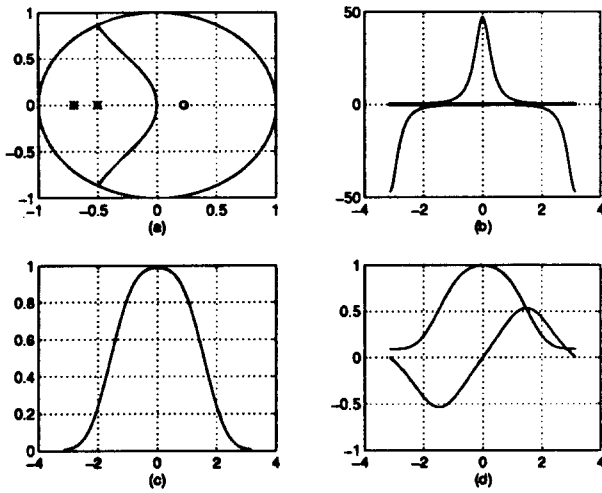


Figure 1. (a) $*$ = 2 Specified Zeros, o = 1 Supplemental Zero. (b) R_1, F, R_2 . (c) Nonnegative Frequency Response of Interpolatory Filter G . (d) Complex Frequency Response of CQF P .

2. FILTER DESIGN

Using the notation in the previous section define filters $B := B_1P_1 - B_2P_2$ and $D := 2P_1P_2$. Therefore

$$R_1 = \frac{B-1}{D}; \quad R_2 = \frac{B+1}{D} \quad (8)$$

and (7) is equivalent to

$$\max_{z \in T} |B(z) - F(z)D(z)| \leq 1. \quad (9)$$

Clearly if F has minimal degree then Q_1 and Q_2 will have minimal degrees. If D has no roots on the unit circle then the minimal degree F that satisfies (6) may be computed (1) by using the standard Remez exchange algorithm if D and B have real coefficients [9], (2) by using a multiple exchange algorithm if D or B has complex coefficients [10], [11]. If D has roots on the unit circle modifications are required.

Definition The set Λ is called *admissible* if it satisfies (3) and there exists $P \in Q(\Lambda)$ having degree $\leq 2n - 1$.

Assume Λ satisfies (2), construct P_1, P_2 as above, and (uniquely) construct minimal length B_1 and B_2 as in [8] so that the filter B has average value 0 over T . The following result is obvious:

Result 3 Λ is admissible if and only if any of the following equivalent conditions hold:

1. $\max_{z \in T} R_1(z) \leq 0 \leq \min_{z \in T} R_2(z)$,
2. the choice $F = 0$ satisfies (7),
3. $|B(z)| \leq 1, z \in T$,
4. the choice $F = 0$ satisfies (9),
5. $\mathcal{u}(\Lambda)$ is admissible.

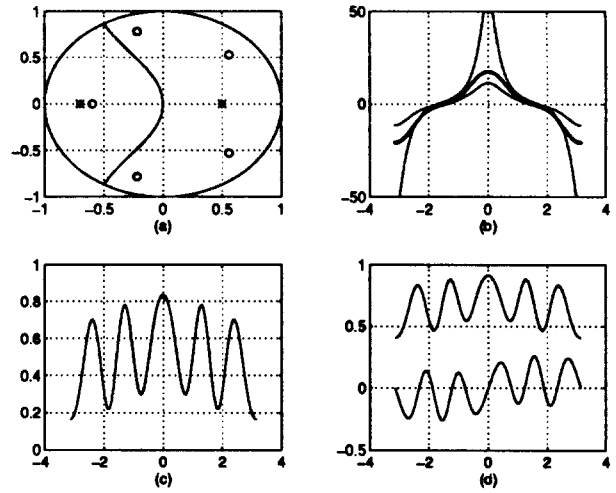


Figure 2. (a) $*$ = 2 Specified Zeros, o = 5 Supplemental Zeros. (b) R_1, F, R_2 . (c) Nonnegative Frequency Response of Interpolatory Filter G . (d) Complex Frequency Response of CQF P .

We consider conditions on Λ that ensure admissibility. From the last condition, we may assume Λ is a subset of the unit disc. The zeros of P that are not in Λ are called *supplemental zeros* of Λ . Define a region $\mathcal{A} \subset C \setminus \{0\}$ by

$$\mathcal{A} := \left\{ x + iy : r^2 := x^2 + y^2 \leq 1; x < -\frac{r^2}{1+r^2} \right\} \quad (10)$$

Result 4 If $n = 1$, Λ is admissible. If $n = 2$ and $\Lambda = \{\lambda, \bar{\lambda}\}$ is a subset of the unit disc then Λ is admissible iff $\Lambda \subset \mathcal{A}$ or $-\Lambda \subset \mathcal{A}$.

Proof The fact $P(z) := (1 + |\lambda|^2)^{-1}(z - \lambda)$ is a CQF for any nonzero λ proves the first statement. Assume $\Lambda = \{\lambda_1, \lambda_2\}$ with $|\lambda_k| \leq 1$ and $\Lambda = \bar{\Lambda}$. Clearly Λ is admissible iff there exists λ_3 with $|\lambda_3| \leq 1$ and $\alpha > 0$ such that $P(z) := \alpha(z - \lambda_1)(z - \lambda_2)(z - \lambda_3)$ satisfies (1). This is equivalent to $\lambda_1 + \bar{\lambda}_1^{-1} + \lambda_2 + \bar{\lambda}_2^{-1} + \lambda_3 + \bar{\lambda}_3^{-1} = 0$. Clearly, there exists λ_3 satisfying this condition if and only if $|\lambda_1 + \bar{\lambda}_1^{-1} + \lambda_2 + \bar{\lambda}_2^{-1}| \geq 2$. If $\lambda_1 = re^{i\theta}$ and $\lambda_2 = re^{-i\theta}$ then this condition becomes $|2(r + r^{-1}) \cos \theta| \geq 2$ which concludes the proof.

Admissibility Conjecture If $\Lambda \subset \mathcal{Z}$ then Λ is admissible.

Figures 1-4 illustrate the design of minimal degree $P \in Q(\Lambda)$ with

$$\Lambda = \{-.7, -.5\}, \{-.7, .5\}, \{-.2 + .7i, -.2 - .7i\}, \{-.6 + .8i, -.6 - .8i, -.2 + .3i, -.2 - .3i, -.5\}$$

respectively. All sets satisfy $\Lambda = \bar{\Lambda}$ and the filters B, D, G, P have real coefficients. The upper left (a) shows the boundary of the unit disc and the set \mathcal{S} , the specified roots by $*$, and the supplemental roots by o . The upper

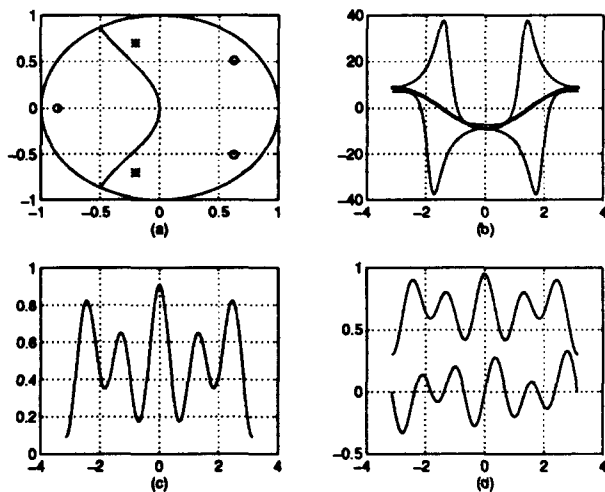


Figure 3. (a) * = 2 Specified Zeros, o = 3 Supplemental Zeros. (b) R_1 , F , R_2 . (c) Nonnegative Frequency Response of Interpolatory Filter G . (d) Complex Frequency Response of CQF C .

right (b) plots $R_1 \leq F \leq R_2$ where the smallest length filter F satisfying the inequalities was computed using the MATLAB Remez algorithm. The lower left (c) plots the frequency response of the interpolatory filter $G = P_1 W$ and the lower right (d) plots the real (upper) and imaginary (lower) components of the frequency response of P . In these cases P is the unique minimal phase factor of G and the zeros of P are the union of Λ and the set of supplemental zeros. The fact Λ in (Fig. 1) is admissible while Λ (Fig. 2) and (Fig. 3) are not is implied by Result 4. The fact Λ in (Fig. 4) is admissible supports the Admissibility Conjecture. All the results illustrate the equivalence of the conditions in Result 3.

3. PROCESSING TRANSIENT SIGNALS

Let P be a CQF whose coefficient sequence p is supported on $\{0, 1, \dots, 2m - 1\}$ and define Q , its twin CQF, to have coefficient sequence q

$$q_n := (-1)^n \bar{p}_{2m-1-n}, \quad n \in \mathbb{Z}. \quad (11)$$

For any sequence s , define sequences $T_p s$ and $T_q s$ by

$$(T_p s)_k := \sqrt{2} \sum_{n \in \mathbb{Z}} \bar{p}_{n-2k} s_n, \quad k \in \mathbb{Z} \quad (12)$$

$$(T_q s)_k := \sqrt{2} \sum_{n \in \mathbb{Z}} \bar{q}_{n-2k} s_n, \quad k \in \mathbb{Z}. \quad (13)$$

Since the sequence q is also a CQF the mapping $s \rightarrow (T_p s, T_q s)$ is unitary (lossless) since for any finitely supported sequence s

$$\sum_{k \in \mathbb{Z}} |s_k|^2 = \sum_{k \in \mathbb{Z}} |(T_p s)_k|^2 + \sum_{k \in \mathbb{Z}} |(T_q s)_k|^2.$$

Therefore, this mapping has an inverse given by its adjoint, and hence s can be exactly reconstructed from $T_p s$ and $T_q s$

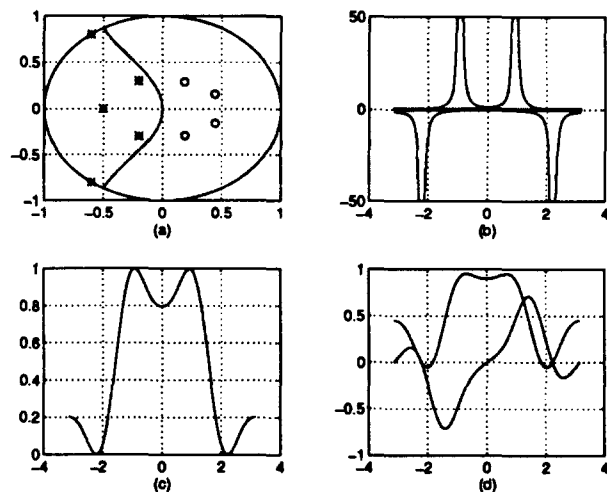


Figure 4. (a) * = 5 Specified Zeros, o = 4 Supplemental Zeros. (b) R_1 , F , R_2 . (c) Nonnegative Frequency Response of Interpolatory Filter G . (d) Complex Frequency Response of CQF C .

by the formula

$$s_k = \sqrt{2} \sum_{n \in \mathbb{Z}} p_{k-2n} (T_p s)_n + \sqrt{2} \sum_{n \in \mathbb{Z}} q_{k-2n} (T_q s)_n, \quad k \in \mathbb{Z}.$$

The design of a CQF with prescribed zeros has potential applications to coding acoustic transient signals. As explained in [8], the mechanical theory of vibrations [12] implies such signals are comprised of linear combinations of functions having the form

$$s_k^j := k^j \lambda^k \quad (14)$$

restricted to half intervals of integers $\mathbb{Z} \cap [a, \infty]$. Furthermore, as shown in [8], if p is the coefficient sequence of a CQF P having $\bar{\lambda}$ as a zero of multiplicity $j + 1$, then

$$(T_h s^j)_k = 0 \quad (15)$$

if $2k$ is sufficiently far from the onset a of the transient.

Figure 5 illustrates coding transient signals using CQF's designed to have specified zeros. Part (a) shows a conjugate pair $\Lambda = \{\lambda, \bar{\lambda}\}$ of specified zeros and 1 supplemental zero for a $P \in Q(\Lambda)$ having degree three. Part (b) shows the real signal s defined by

$$s_k := \lambda^k \chi_{[1, \text{inf}]}(k) + \bar{\lambda}^k \chi_{[1, \text{inf}]}(k), \quad (16)$$

which consists of a sum of functions described by (14). Parts (c) and (d) show the signals $T_p s$ and $T_q s$, where s is the twin CQF sequence defined in (11), and T_p, T_q are defined by (12), (13) respectively. Note the subband coding has essentially compressed the signal s into half the number of samples.

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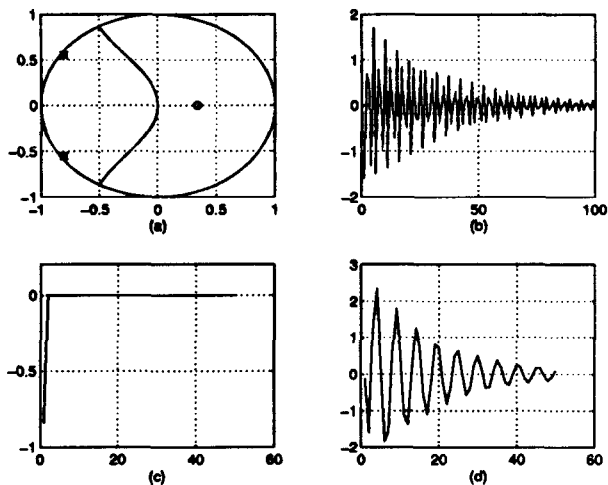


Figure 5. (a) zeros of filter (b) transient acoustic signal corresponding to these zeros, (c) decomposition of suppression band (d) decomposition of passing band.

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