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t-Fault t/2-Step Sequentially Diagnosable Systems

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Abstract—Fault diagnosis is considered for systems comprising n units u_0, u_1, \dots, u_{n-1} in which u_i tests u_{i+1} and u_{i+2} . A necessary and sufficient condition for the diagnosability of t faults in $t/2$ or $t/2 + 1/2$ steps is stated, the condition taking the form of an inequality relating n and t .

I. INTRODUCTION

Since the introduction of graph theoretic models for system level diagnosis by Preparata *et al.* [1], a number of other models have been developed (see [2] for references) and analyzed. In this letter we will use Preparata's model for a special class of systems and state necessary and sufficient conditions for fault diagnosability on the number of nodes n as a function of the number of faults t in the system. Full proofs will be given elsewhere.

The system is composed of n units u_0, u_1, \dots, u_{n-1} where each unit is capable of testing other units. The system is represented by a directed graph where nodes correspond to units and arcs to test links. The outcome of the test applied by u_i to u_j is denoted by a binary variable a_{ij} where

$$\begin{aligned} a_{ij} &= 1, \text{ if } u_i \text{ is fault free and } u_j \text{ is faulty} \\ a_{ij} &= 0, \text{ if } u_i \text{ is fault free and } u_j \text{ is fault free} \\ a_{ij} &= d, \text{ if } u_i \text{ is faulty, where } d \in \{0, 1\}. \end{aligned}$$

Necessary and sufficient conditions for $D_{\delta t}$ systems [1] to be one step diagnosable and single loop systems [1], [3] to be sequentially diagnosable are known. In this correspondence we give necessary and sufficient conditions for $D_{\delta 2}$ systems to be $t/2$ -step or $(t+1)/2$ -step sequentially diagnosable. In the following section, the model is defined and in Section III the results are presented.

II. MODEL

A system with n -units u_0, u_1, \dots, u_{n-1} is said to be $D_{\delta 2}$ if its graph has the following property: there is a test link from u_i to u_j if and only if $j - i = \delta m \pmod n$ for $m = 1, 2$. It can easily be shown that if δ and n are relatively prime, a $D_{\delta 2}$ system is isomorphic to a D_{12} system (e.g., see [1], [4]). A D_{12} system with $n = 9$ is shown in Fig. 1. Therefore, we shall only study D_{12} systems. In a D_{12} system, certain fault patterns have the property that at most two faults can be diagnosed in one step. (Consider, for example, a fault pattern in which precisely $u_i, u_{i+1}, \dots, u_{i+k}$ are faulty. Even if u_i and u_{i+1} can be diagnosed as faulty, there is no way, if $a_{\alpha\beta} = 0$ for $\alpha \in [i, i+k]$, that any of u_{i+2}, \dots, u_{i+k} could be diagnosed as being fault free or faulty.)

It follows that to diagnose t faults, we might require at least $\lceil t/2 \rceil$ steps (where $\lceil x \rceil$ denotes the smallest integer $\geq x$), it being understood that at each step, two faulty units are diagnosed and replaced by fault-free units. This observation gives rise to the following problem statement.

Find the smallest number of units n such that a D_{12} design of a system with n units is sequentially $\lceil t/2 \rceil$ -step t -fault diagnosable.

Clearly such a system will be optimal with respect to the number of links and number of units n .

III. RESULTS

For a given value of t , we define

$$n_1 = \lceil t/4 \rceil (t - 2\lceil t/4 \rceil) + t + 3$$

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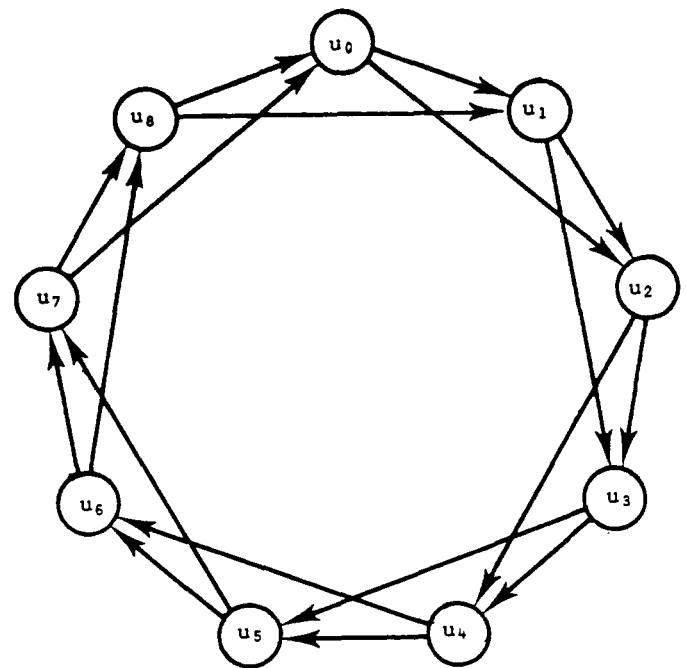


Fig. 1. A D_{12} system with $n = 9$.

TABLE I
MINIMUM VALUE OF n FOR SINGLE LOOP AND D_{12} SYSTEMS

t	Single loop system	D_{12} system		
		n_1	n_2	$n \geq \max(n_1, n_2)$
2	5	5	5	5
3	7	7	6	7
4	10	9	9	9
5	13	10	11	11
6	17	13	13	13
7	21	16	15	16
8	26	19	19	19
9	31	21	22	22
10	37	25	25	25

and

$$n_2 = \lceil t/4 \rceil (t - 2\lceil t/4 \rceil) + t + 3$$

where $\lceil x \rceil$ is the largest integer $\leq x$.

The following theorem gives the necessary and sufficient conditions for a D_{12} system to be sequentially $\lceil t/2 \rceil$ -step t -fault diagnosable. The proof of the theorem is quite long and complex and is not included here.

Theorem 1: A D_{12} system with n units is sequentially $\lceil t/2 \rceil$ -step t -fault diagnosable, for $t \geq 2$, if and only if

$$n \geq \max(n_1, n_2).$$

In view of our earlier comment we can say that the result of Theorem 1 also applied to $D_{\delta 2}$ systems with coprime δ and n and also to systems which may contain $D_{\delta 2}$ systems as subsystems. Table I lists the minimum value of n for different $t (\leq 10)$ for D_{12} and single loop systems (where t -step diagnosis is being considered [1]).

It is evident from Table I that for small systems ($n \leq 13$), n meets the absolute necessary minimum bound (i.e. $n \geq 2t + 1$, see [1]) for t -fault diagnosability.

IV. CONCLUSIONS

The main result as stated in the theorem requires of itself little comment. What, though, of extensions? An obvious direction to extend

the ideas is to consider conditions for $\lfloor t/m \rfloor$ -step diagnosability for arbitrary m in $[1, t]$. The authors have developed a necessity condition which they conjecture to be sufficient. They have also studied 2-step diagnosability, and are able to strengthen in a minor way a result of [4]. These results will be reported separately.

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Correction to "Computerized Geophysical Tomography"

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In the above article,¹ equations (12) and (18) contain misprints. The correct equations are

$$\Delta f_{ijk}^A = \frac{(\Delta s_{ijk})^w \Delta g_k}{\sum_i \sum_j (\Delta s_{ijk})(\Delta s_{ijk})^w} \quad (12)$$

and

$$\Delta f_{ij}^A = \frac{\sum_{k=1}^K \Delta f_{ijk}^A / N_k^4}{\sum_{k=1}^K \text{sgn}(\Delta s_{ijk}) / N_k^4} \quad (18)$$

where N_k is the number of picture cells intercepted by the k th ray. In addition, equation (11) is not labeled.

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¹K. A. Dines and R. J. Lytle, *Proc. IEEE*, vol. 67, pp. 1065-1073, July 1979.

On True 3-D Object Reconstruction from Line Integrals

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Abstract—An algorithm for true three-dimensional reconstruction from line integrals (projections onto planes) is discussed and a three-dimensional filter equivalence theorem is formulated. A Radon-type equation is obtained, and the global character of the reconstruction is established. The relation of the algorithm to the conventional slice by slice reconstruction is examined.

In a recent communication Nalcioglu and Cho [1] reported on a 3-D analog of the 2-D Fourier synthesis operator for object reconstruction from its line integrals. The purpose of this communication is to relate one of their results to a filter equivalency theorem in the Fourier synthesis and the filtered summation-image methods. The global nature of the reconstruction algorithm, as well as the redundancy in the experimental data required by the algorithm, are discussed.

THE FILTER EQUIVALENCY THEOREM

Barrett *et al.* [2], [3] proved a theorem on the identity of the functional forms of the one- and two-dimensional filters in 2-D object

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reconstruction by the Fourier-synthesis and filtered summation-image methods. Their corresponding equations are

$$\hat{f}_1(x, y) = \beta_2 \{ F_1^{-1} [H_1(|\rho|) \cdot F_1 \{ g \}] \} \quad (1)$$

$$\hat{f}_2(x, y) = F_2^{-1} [H_2(\rho) \cdot F [\beta_2 \{ g \}]] \quad (2)$$

Here $F_1[]$ and $F_2[]$ indicate one- and two-dimensional Fourier transforms, $\beta_2 \{ \} = \int_0^\pi d\theta \{ \}$ is the two-dimensional linear superposition operator that yields the two-dimensional summation image, g is a line-integral of a two-dimensional object, and ρ is the spatial frequency in (1) and the length of the radius-vector of a point in the Fourier plane in (2). Barrett's theorem states that if the image estimates \hat{f}_1 and \hat{f}_2 are equal, then

$$H_1(|\rho|) = H_2(\rho) \quad (3)$$

In the special case of a purely mathematical reconstruction, i.e., without additional filtering, $H_1 = |\rho|$ and $H_2 = \rho$.

The equations analogous to (1) and (2) in the 3-D case are

$$\hat{f}_1(x, y, z) = \beta_3 \{ F_2^{-1} [K_2(\rho_1, \rho_2) \cdot F_2 \{ h \}] \} \quad (4)$$

$$\hat{f}_2(x, y, z) = F_3^{-1} [K_3(\rho) \cdot F_3 \{ \beta_3 \{ h \} \}] \quad (5)$$

Here $F_3[]$ indicates the three-dimensional Fourier transform, $\beta_3 \{ \} = \int_0^\pi \int_0^\pi \sin \theta d\theta d\varphi \{ \}$ is the three-dimensional linear superposition operator that yields the three-dimensional summation image, h is a line-integral over a three-dimensional object along a line parallel to a direction specified by the angles θ and φ in a spherical coordinate system, ρ_1 and ρ_2 in (4) are the spatial frequency components in the Fourier plane, and ρ in (5) is the length of the radius-vector of a point in the 3-D Fourier space.

By a rather lengthy derivation for the algorithms represented by (4) and (5) it can be shown that

$$K_2(\rho_1, \rho_2) = K_3(\sqrt{\rho_1^2 + \rho_2^2}) \quad (6)$$

provided that the image estimates are equal.

Equation 6 represents a generalization of Barrett's theorem to a 3-D object reconstruction from its line integrals. For a purely mathematical reconstruction $K_3 = \rho/\pi$ and $K_2 = \sqrt{\rho_1^2 + \rho_2^2}/\pi$. The filter equivalency theorem reflects on the relationship between the Fourier-synthesis and the filtered summation-image methods and can be used for the development of specific filtering procedures, for example, those based on regularization theory for ill-posed problems [4].

DISCUSSION

When $K_2 = \sqrt{\rho_1^2 + \rho_2^2}/\pi$, equation (4) can be written explicitly in a form that corresponds to (9) given by Nalcioglu and Cho [1].

$$f(x, y, z) = 1/\pi \int_0^\pi \int_0^\pi \left\{ \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} \sqrt{\rho_1^2 + \rho_2^2} \tilde{h}(\rho_1, \rho_2; \theta, \varphi) \cdot \exp [2\pi i(x'\rho_1 + y'\rho_2)] d\rho_1 d\rho_2 \right\} \sin \theta d\theta d\varphi \quad (7)$$

where

$$\tilde{h}(\rho_1, \rho_2; \theta, \varphi) = \iint_{-\infty}^{\infty} h(x', y'; \theta, \varphi) \exp [-2\pi i(x'\rho_1 + y'\rho_2)] dx' dy' \quad (8)$$

Here x' and y' are rectangular coordinates in the plane onto which all parallel line integrals are projected. The inverse Fourier transform of $\sqrt{\rho_1^2 + \rho_2^2}$ is $-1/4\pi^2(x'^2 + y'^2)^{-3/2}$ (see [5, p. 197]). It is a generalized function and is defined as a certain functional. As a part of the definition of this generalized function the singular integral related to it has to be regularized by removing the singularity. (See [5] p. 10, 45, and further.) Substituting the inverse Fourier transform of $\sqrt{\rho_1^2 + \rho_2^2}$ into (7) and using the convolution theorem we get the following intermedi-